IP Office 4.2

IP Office Product Description

15-601011 Issue 18.b.- (08 July 2008)

© 2008 AVAYA All Rights Reserved.

Notice

While reasonable efforts were made to ensure that the information in this document was complete and accurate at the time of printing, Avaya Inc. can assume no liability for any errors. Changes and corrections to the information in this document may be incorporated in future releases.

Documentation Disclaimer

Avaya Inc. is not responsible for any modifications, additions, or deletions to the original published version of this documentation unless such modifications, additions, or deletions were performed by Avaya.

Link Disclaimer

Avaya Inc. is not responsible for the contents or reliability of any linked Web sites referenced elsewhere within this Documentation, and Avaya does not necessarily endorse the products, services, or information described or offered within them. We cannot guarantee that these links will work all of the time and we have no control over the availability of the linked pages.

License

USE OR INSTALLATION OF THE PRODUCT INDICATES THE END USER'S ACCEPTANCE OF THE TERMS SET FORTH HEREIN AND THE GENERAL LICENSE TERMS AVAILABLE ON THE AVAYA WEBSITE AT http://support.avaya.com/LicenseInfo/ ("GENERAL LICENSE TERMS"). IF YOU DO NOT WISH TO BE BOUND BY THESE TERMS, YOU MUST RETURN THE PRODUCT(S) TO THE POINT OF PURCHASE WITHIN TEN (10) DAYS OF DELIVERY FOR A REFUND OR CREDIT.

Avaya grants End User a license within the scope of the license types described below. The applicable number of licenses and units of capacity for which the license is granted will be one (1), unless a different number of licenses or units of capacity is specified in the Documentation or other materials available to End User. "Designated Processor" means a single stand-alone computing device. "Server" means a Designated Processor that hosts a software application to be accessed by multiple users. "Software" means the computer programs in object code, originally licensed by Avaya and ultimately utilized by End User, whether as stand-alone Products or pre-installed on Hardware. "Hardware" means the standard hardware Products, originally sold by Avaya and ultimately utilized by End User.

License Type(s): Designated System(s) License (DS).

End User may install and use each copy of the Software on only one Designated Processor, unless a different number of Designated Processors is indicated in the Documentation or other materials available to End User. Avaya may require the Designated Processor(s) to be identified by type, serial number, feature key, location or other specific designation, or to be provided by End User to Avaya through electronic means established by Avaya specifically for this purpose.

Copyright

Except where expressly stated otherwise, the Product is protected by copyright and other laws respecting proprietary rights. Unauthorized reproduction, transfer, and or use can be a criminal, as well as a civil, offense under the applicable law.

Third-Party Components

Certain software programs or portions thereof included in the Product may contain software distributed under third party agreements ("Third Party Components"), which may contain terms that expand or limit rights to use certain portions of the Product ("Third Party Terms"). Information identifying Third Party Components and the Third Party Terms that apply to them is available on Avaya's web site at: http://support.avaya.com/ThirdPartyLicense/

Avaya Fraud Intervention

If you suspect that you are being victimized by toll fraud and you need technical assistance or support, call Technical Service Center Toll Fraud Intervention Hotline at +1-800-643-2353 for the United States and Canada. Suspected security vulnerabilities with Avaya Products should be reported to Avaya by sending mail to: securityalerts@avaya.com.

For additional support telephone numbers, see the Avaya Support web site (http://www.avaya.com/support).

Trademarks

Avaya and the Avaya logo are registered trademarks of Avaya Inc. in the United States of America and other jurisdictions. Unless otherwise provided in this document, marks identified by "®," "™" and "SM" are registered marks, trademarks and service marks, respectively, of Avaya Inc. All other trademarks are the property of their respective owners.

Documentation information

For the most current versions of documentation, go to the Avaya Support web site (http://www.avaya.com/support) or the IP Office Knowledge Base (http://marketingtools.avaya.com/knowledgebase/).

Avaya Support

Avaya provides a telephone number for you to use to report problems or to ask questions about your contact center. The support telephone number is 1 800 242 2121 in the United States. For additional support telephone numbers, see the Avaya Web site: http://www.avaya.com/support.

Contents

1. Introduction

Avaya IP Office Family 6
What's New in IP Office 4.2 10
Voice Communication Solution Features 18
Data Communication Solution Features 19
Applications Platform Features 20
Management Tools 21
Scalable Platform 21
Telephone Options 22
Application and Feature Licensing 22

2. IP Office Platform

IP Office 500 Control Unit 25
IP500 Cards 27
IP500 Trunk Cards 30
IP Office IP412 Control Unit 32
IP400 Trunk Interface Cards 34
Internal Daughter Cards 36
External Expansion Modules 37

3. Telephones

1700 47
4601 Telephone 55
5402, 5602 SW, 2402, 4602 SW Telephone 56
5410, 5610 SW, 2410, 4610 SW Telephones 58
5420, 5621, 2420, 4621, 4625 Telephones 60
EU24 and EU24 BL Expansion Modules 62
T3 Telephone Range 63
Wireless and Remote Phone Solutions 68
Other Ranges 85
Headsets 96
Summary 97

4. Features

Basic Call Handling 100
Advanced Call Handling 104
Key and Lamp Operation 111
Outbound Call Handling Features 113
Forwarding 115
Avaya Digital and IP Phones 117
Inbound Call Handling 119
Contact Center Features 122
Miscellaneous Features 124
System Short Codes 125

5. IP Telephony

Gateways, Gatekeepers and H.323 - Technology Overview 130
IP Telephony Features 131
Power Options for IP Telephones 132
VoIP FAQ 134

6. Public and Private Voice Networks

Private Circuit Switched Voice Networking 141
Public Voice Networking 142
Session Initiation Protocol (SIP) 146
Packet Based Voice Networking 149
Supplementary Services within IP Networks 151
Small Community Networking 152
Internetworking with Other Avaya Products 154
Common Networking Features 157

7. Data Networking Services

LAN/WAN Services 161
Quality of Service 161
Internet Access 162
Remote Access Features 163
LAN to LAN Routing 163
Data Networking Features 164

8. Phone Manager

Phone Manager Lite 173
Phone Manager Pro 176
Phone Manager PC Softphone (IP Softphone) 179
Phone Manager Feature Summary 180
Phone Manager PC Requirements 181

9. SoftConsole

SoftConsole Options 189
SoftConsole Administration 190
SoftConsole PC Requirements 190

10. Computer Telephony Integration

CTI with IP Office 194
TAPILink Lite (1st Party TAPI Support) 195
TAPILink Pro (3rd Party TAPI Support) 195
Support for Developers 195

11. Voicemail

Voicemail Lite 198
Embedded Voicemail 199
Voicemail Pro 200
Interaction of Voicemail with Email Systems 202
Unified Messaging Service (UMS) 203
Fax Messages 205
Email Reading (Microsoft Exchange only) 206
Networked Messaging 207
Auto Attendant 208
Accessing Database Information within Call Flows (IVR) 209
Using Text To Speech (TTS) Facilities within a Call Flow 210
Visual Basic (VB) Scripting 210
Personal Numbering 211
Extended Personal Greetings 212
Hunt Group Broadcast Messages 213
Personal Distribution Lists 213
Cascaded Out-Calling 214
Campaign Manager 215
Call Recording 216
IP Office ContactStore 217
Centralized Messaging with Avaya Communication Manager 218

© 2008 AVAYA All rights reserved.

Voicemail Feature Comparison 219
PC Requirements 223

12. Audio Conferencing

IP Office Meet-Me Conferencing Solution 229
IP Office Conferencing Capacity 230
IP Office Standard Conferencing Features 231
Conferencing Center 232

13. The Contact Center

SBCC 242
Compact Contact Center 251

14. CRM Integration

Inbound Call Operation 261
Outbound Call Operation 262

15. Management Utilities

IP Office Manager 266
Monitor 269
Simple Network Management Protocol (SNMP) 270
CDR 271
SMDR 272
System Status Application 273

16. Configurations

Country Availability 281
Sample Configurations 282

17. TAPI Functions Supported by IP Office

TAPI 2.1 Functions Supported 286
TAPI 3.0 functions supported 286
Device-Specific Data 287
DevLink Fields 288

18. Technical Specifications

Interfaces 294
Protocols 295

19. Appendix

IP406 V2 Control Unit 299
IP400 WAN3 10/100 302
IP400 So8 Expansion Module 302
Small Office Edition 303

© 2008 AVAYA All rights reserved.

Chapter 1. Introduction

© 2008 AVAYA All rights reserved.

1. Introduction

This document forms no part of a contract, the specification of the Avaya IP Office family is subject to change without notice. Not all components and features documented are available in all territories, refer to your Avaya Representative for further details. This document should be read in conjunction with any issued technical bulletins and/or product offer announcements.

Avaya IP Office Family

The Intelligent Communications solution for small and midsize businesses.

What is IP Office?

A solution for voice and data communications, messaging and customer management. It uses IP technology to deliver more functionality at a lower cost.

How can I use it in my business?

To connect with colleagues and customers… simplify access to information… keep remote workers in touch. To save money through conferencing, networking, time/ call management, Voice over IP and more.

What are my choices?

Does your business have one location? Multiple locations? Are you a branch office of a larger organization? A home office? With IP Office you can choose from a range of models and add capacity, applications and phones, as you need them. Whether you have 2 employees, 200 or more, IP Office is the right choice.

IP Office: Three key things to know

Every small- and medium-size business needs ways to reduce costs and improve the way it operates. Like every business, you're looking to keep all your customers, add new ones and grow at the pace that's right for you. Avaya understands this. With over one hundred years of experience as a leader in communications, we know that the right solution for your business is one that helps you increase profitability, improve productivity and gain competitive advantages.

Get big business communications —at small business prices

Over one million businesses rely on Avaya solutions like Avaya IP Office —the award-winning business communications system that gives growing companies an "all-in-one" solution for telephony, messaging, networking, conferencing, customer management and much more. Growing businesses know they can rely on Avaya for big business capabilities at small business prices—Avaya has an entire division focused on the needs of small- and medium-size businesses. We support extensive research into new technologies and standards, and we make it easy for businesses like yours to acquire our solutions by offering an array of financing options.

See what Avaya can do for you

You need a communications system—every business does. To find one that's right for your business, start with Avaya. With solutions like IP Office, we're revolutionizing how small and medium businesses communicate. Now is the time to see what an Avaya solution can do for your business.

Reduce monthly costs. Now.

IP Office will help you lower the cost of communications, with capabilities like conferencing, making calls over a managed Internet service (Voice over IP) and the "all-in-one" benefits of a converged communications system.

Leave the office. Be accessible.

With easy, flexible options for call/message forwarding and one-number reachability, IP Office keeps everyone in touch. Get the freedom to go where you want and never miss important business calls.

© 2008 AVAYA All rights reserved.

Serve better. Sell more.

IP Office can give you a customer sales and service center designed for your needs and your budget—with all the routing and reporting capabilities you need. Deliver the personal service that builds sales and loyalty.

Get connected.

Talk to your Avaya BusinessPartner. Discuss where you want communications to add value to your business. Learn about the different service and support options that are available. See why thousands of growing businesses rely on the innovative Avaya IP Office solution.

The right choice for you and your business.

How we communicate is a personal choice - it has to match the needs of your business. And your needs change depending on whether your employees are working in the office, at home, or on the road. That's why when you choose IP Office you can also choose from a whole range of communication tools and applications designed to boost productivity. Choose a basic phone or one with all the bells and whistles. Connect our IP phones directly to your office LAN - also use them at home and get all the features you have at the office. Avaya Phone Manager software can turn the screen of your PC into a phone. And our wireless solutions make it easier to roam the office. With all of our IP Office capabilities, our goal is to make your communications simple and cost-effective. Let your Avaya BusinessPartner put together a selection of tools and applications that's right for you.

Fine-tuning performance.

How many calls are you handling an hour, a day? What are your peak calling periods? How many calls typically turn into sales? Avaya IP Office reporting capabilities can help you measure and manage your availability and response to customers.

Day-to-day administration.

Once your system is up and running you will benefit from the menu-driven administration tools that simplify day-to-day tasks, such as updating directories and moving phone extensions.

Getting started.

Is your communications network ready for IP Office? We'll make sure. Avaya has created a whole set of assessment and automated configuration tools to make sure that when your system is installed it's ready to meet your needs starting Day One.

Keeping ongoing management simple.

Concerned about needing extra resources to administer a system as powerful as IP Office? There's no need for worry. IP Office comes with a whole set of menu-driven tools to keep ongoing management simple.

Does my current phone system give my business what it needs?

If it is based on old technology, probably not. Your competitors will react faster and appear more professional with the latest in communications software. IP Office delivers the capabilities that allow you to keep up with or overtake the competition.

Do I need to understand the technology to implement it?

No. IP Office is designed specifically to give you more functionality without making more demands on your resources. Rely on your certified Avaya BusinessPartner for support before, during and after your purchase. We'll take care of you so you don't have to worry.

Do I need to spend a lot?

Not at all. You have choices based on your budget needs. Easy leasing or financing plans not only make this affordable; they help you quickly cut monthly expenses immediately. And you only have to buy/lease what you need, when you need it.

© 2008 AVAYA All rights reserved.

Is IP technology so new that it's not reliable?

With over 130,000 systems deployed worldwide (Avaya is #1 in Enterprise and IP Telephony shipments - source Dell'Oro and Synergy Research, Q4 2007), Avaya IP Office has the track record businesses like yours can rely on . Aside from receiving the Product of the Year award by Internet Telephony magazine and being named Best in Test by Miercom in 2004, customers like you are saving money and boosting productivity. Many are managing the system themselves via menu-driven tools.

I have old systems but am adding an office. Should I consider the new technology?

Not only would this be a way for you to experience the rich functionality of the latest communications applications, but we may be able to network with your existing equipment, as well as provide a gradual migration plan for your other locations.

How quickly can I get up and running?

Just say "when"—an authorized Avaya BusinessPartner can tailor a solution to your needs and your budget. By saving you money and helping you grow, IP Office repays your investment and lets you reallocate resources to other business priorities.

Lowering long distance costs.

Routing phone calls over IP lines—Voice over IP—is growing in popularity. Particularly in the case of international calls, VoIP generates significant savings. If your company is already linking multiple offices using high-speed lines, the VoIP capabilities in IP Office make it possible to route voice calls over the existing infrastructure, providing another way to lower costs and leverage your investment. However you do it, the VoIP capabilities of IP Office are a way to put money back in your pocket.

Eliminating conferencing fees.

For connecting with partners, suppliers and dispersed employees, conference calls keep people working together and keep travel costs down. Many companies rely on third party teleconferencing services and pay a price for the convenience. This is particularly true—and irritating—if a call that's scheduled doesn't happen: you still pay the fee.
Now there's an alternative that will save you money. With Avaya IP Office, your organization can have its own private, secure conference bridge and entirely eliminate fees to third party providers.

Supporting multiple offices/remote workers

When employees can't get to the office (because of storms, medical issues or other reasons) but can still work productively at home, your business benefits. IP Office Phone Manager lets you turn any PC into a phone, making it easy and productive to work anywhere. And the ability to network phone systems and share messaging systems between offices reduces up front investment and drives long-term productivity.
And keep in mind…

IP Office delivers a whole range of capabilities. Only you can put a number on the value that many of these capabilities will have for your business.
Examples:

- Having calls automatically routed to a cell phone or other location, so important customers can get through to the right person in real-time
- Being able to operate as a 24/7 business, without a 24/7 staff
- Using your communications to quickly identify when your top customers call.

© 2008 AVAYA All rights reserved.

How IP Office is benefiting businesses today.

- **More room for sales**
  With IP Office, a leading provider of commercial food service equipment now handles 50% more calls per day, without extra staff and without sacrificing the personal service it knows is the key to sales.
- **At the head of the class**
  By relying on IP Office to connect nearly 50 buildings, a public school system saved thousands of dollars on inter-office calls and simplified communications.
- **Lowering global costs**
  By using IP Office to hold teleconferences and make phone calls across the IP network, a strategic consulting firm is saving up to $30,000 per year.

The right model for your business with several models to choose from, there's an IP Office to meet your needs. Ready to grow Capacities: 2-360 extensions; up to 192 analog lines; 192/240 T1/E1 lines; 128 VoIP trunk lines.

Call handling and messaging.

Get 24-hour support for callers/customers without a 24-hour staff. IP Office has a range of messaging, auto attendant and Interactive Voice Response (IVR) capabilities. Integrate messaging and advanced call handling into your customer service operations. Handle voice mail and email in a single mailbox.

Communication with customers.

Set up a formal or informal customer service center. Integrate your customer data base into your call handling. Manage the quality of your customer interactions.

Work anywhere.

Give your employees all the communications capabilities they have at the office whether they are working from home, a hotel or a remote office.

A complete conferencing solution.

Don’t pay any more fees to outside conferencing service providers. Get Web and audio-based conferencing that are easy to set up and use.

Secure converged communications.

Use IP Office as a secure router with a built-in firewall/VPN. Route voice calls over a managed Internet service (VoIP) and pocket the savings. Simple administration Windows-based, menu-driven tools cut the time and expense of administration.

© 2008 AVAYA All rights reserved.

What's New in IP Office 4.2

For those already familiar with IP Office, this section lists the new features introduced in IP Office 4.2. This is not an exhaustive list; however, it covers the major changes that are aimed at improving product manageability, flexibility and end user mobility.

Hardware Support

IP Office Control Unit Support

- IP Office 4.2 is supported on the IP Office 500 and IP412 as well as the Small Office Edition and IP406 V2 (no longer sold).
- IP Office 4.2 is not supported on the IP403 and IP406 V1.

Note: Not all of the features in IP Office 4.2 are supported on all platforms - please see each feature's description for details.

1700 Series IP Telephone Support

IP Office 4.2 supports the Avaya IP 1700 series telephones - models 1703, 1708 and 1716, together with the BM32 button module. The Avaya 1700 series form a family of cost-effective IP telephones that deliver familiar features at an attractive price point for customers with basic communications needs. Designed as a stand-alone portfolio these telephones may be intermixed with other Avaya IP telephones. Combining the features of traditional telephones and the latest in telephone user experience, Avaya IP 1700 series telephones provide critical capabilities not often found in competitively priced models.

Legacy Telephone Support

The following telephones are not supported by IP Office 4.2. They may function but have not been tested with IP Office 4.2 and any faults reported with IP Office 4.2 will not be fixed.

- 20DT Analog DECT used with IP Office Analog DECT and Compact DECT
- 4606, 4612 and 4624 IP phones
- TransTalk 9040

So/To Interfaces for IP500 BRI

IP Office 4.2 permits circuits on the IP500 Dual and Quad BRI trunk interfaces (daughter cards) to operate as So interfaces in the same way as ports on the S08 Expansion Module. IP Office provides clocking, and runs the ETSI Protocol as network rather than TE equipment.

The IP500 BRI card in So mode supports both point-to-point and point-to-multipoint connections. A maximum of 10 terminal endpoints identifiers (TEIs) are supported on each bus. There is no hardware modification to the cards and as a result there will be no 'signalling power'. This feature does require crossover cables and line termination (if not provided by the terminal equipment). A suitable converter cable is available from Avaya (material code 700458649).

Voice over IP Enhancements

IP Office 4.2 introduces changes to allow easier installation of Avaya 1700, 4600 and 5600 Series phones as well as T3 IP and IP DECT phones, with minimal user intervention and to allow greater configurability for more complex network environments.

DHCP Enhancements (IP500 only)

In IP Office 4.2, the embedded DHCP server can now support 8 non-contiguous pools of address ranges, each fully configurable into different subnets. The server can be configured to only serve Avaya VoIP telephone clients. It will also offer consistent IP addresses to clients after reboots.

Server Enhancements (IP500 only)

In IP Office 4.2 the total number of clients that can be served by all embedded servers (DHCP, TFTP/HTTP, RAS/H323) has been increased to 50.

Embedded File Management (IP500 and IP406 V2 only)

IP Office 4.2 provides a new user interface within Manager to list the directory content of the removable Compact Flash card, as well as upload, download and delete files.

IP Phone Forced Re-Registration

This new option within the IP Office 4.2 System Status Application (SSA) permits the remote re-registration of individual or groups of Avaya IP telephones for firmware upgrade and maintenance purposes.

© 2008 AVAYA All rights reserved.

User and Extension Auto Creation Enhancements

The auto creation feature for IP Phone users and extensions has been made consistent across phone types in IP Office 4.2.

Mobility Enhancements

Mobile Call Control

In IP Office 4.2, when a call to a user is delivered to an external number, due to External/Mobile Twinning, the user will be able to dial a DTMF sequence **(star-star) to put the call on hold and receive IP Office system dial tone. The user will then be able to perform supervised and un-supervised transfers, shuttle (switch between held calls), and conference (Conference add and transfer to a meet-me conference).

This feature is only supported through the following trunk types on an IP Office 500:

- IP500 PRI-U (single or dual) daughter cards
- IP500 Dual or Quad BRI daughter cards
- SIP trunks (RFC 2833)
- This feature requires one instance of the Mobility License as described below for each configured user.

Avaya one-X Mobile Client Support

IP Office 4.2 supports release 4.0 of the Avaya one-X Mobile Client "Single Mode" operation, in conjunction with mobile technology Symbian Single-Mode Version 4.0 and Windows Mobile 6 technology. This provides a graphical interface for call control and allows a one-number service for both incoming and outgoing calls.

Avaya one-X Mobile builds on the functionality of the IP Office 4.2 by providing a family of mobile clients designed to further enhance the productivity of mobile workers. The family of Avaya one-X Mobile clients provides quick access to office phone functionality via an intuitive graphical user interface. The clients support devices ranging from high end smart phones to lower end feature phones to support most business environments, independent of the mobile devices deployed.

This feature is only supported through the following trunk types on an IP Office 500:

- IP500 PRI-U (single or dual) daughter cards
- IP500 Dual or Quad BRI daughter cards
- SIP trunks (RFC 2833)
- This feature requires one instance of the Mobility License as described below for each configured user.

Mobile Twinning license now known as Mobility license

In IP Office 4.2 the existing Mobile Twinning license has been renamed to Mobility license. The license now enables the use of Mobile Call Control and one-X Mobile client features as well as Mobile Twinning. An instance of the license will be consumed for every user who has Mobility Features checked in IP Office Manager.

Mobile Twinning for Logged Out Users

In releases prior to IP Office 4.2, a user had to be logged in to an extension to use Mobile Twinning. IP Office 4.2 enables the user to logout from an extension and maintain a logical presence on the system through Mobile Twinning. Logged Out includes the extension error state where the user's association to an extension is broken due to the extension being unplugged or a network fault.

© 2008 AVAYA All rights reserved.

Management Enhancements

Static NAT

This new feature is designed primarily for dial-up diagnostic requirements and maintenance abilities. For example this could be utilised to give access to VoiceMail Pro servers.

IP Office Firewall settings now support Static NAT, with the ability to `pair' external to internal IP addressing. Up to 64 such pairs can be created programming via the Manager.

Static NAT allows the following applications though the Static NAT translations: -

- PC-Anywhere
- Windows Terminal Server
- Ping and Tracert
- Manager
- Sysmon
- Upgrade Wizard
- SSA
- VM Pro Client

System Status Application enhancements

In IP Office 4.2, System Status Application (SSA) has a new real-time configuration alarms screen which shows the following new alarms:

- Duplicate Numbers in Small Community Networking dial plan
- "No routing for Line" - Incoming Line Group has no associated Incoming Call Route
- Call Routed to a non-existent Start Point in VoiceMail Pro
- SSA displays the following new resource availability alarms in IP Office 4.2:
  - Attempt to use a feature for which no license is installed
  - Voicemail storage full or nearly full
- The following new screens are provided in SSA in IP Office 4.2:
  - Control Unit Audit - details the last 15 accesses from Manager
  - Feature Key and License Status
  - Directory, including Users on the SCN
  - Real time Voicemail port status
  - Voicemail Summary/Detail information
  - Real time Mailbox status
  - IP Route and Service status
  - Tunnel Status

The new option "IP Phone Forced Re-Registration" within SSA permits the remote re-registration of individual or groups of Avaya IP telephones for firmware upgrade and maintenance purposes.

© 2008 AVAYA All rights reserved.

Manager Enhancements

All Multi Line Options are now grouped together under a new tab under User/Telephony.

Manager now displays the system/default values for No Answer Time and Ring Delay.

DSS Status and Beep on Listen have been moved to the System/Telephony Tab.

The Conferencing Tone checkbox has been replaced with two options: "Entry and Exit tones" and "Repeating Tone" are now found in the new Tones and Music options section under System/Telephony.

IP Office 4.2 Manager can now limit the number of BAK files to keep - under Manager Preferences.

Embedded Voicemail Enhancements

Voicemail to Email (IP500 and IP406 V2 only)

IP Office 4.2 supports the ability to send an email containing either notification or the complete voicemail message from the Embedded Voicemail when using IP406 V2 and IP500 units only.

Cantonese Voicemail Prompts (IP500 and IP406 V2 only)

When using the Hong Kong locale, Embedded Voicemail prompts are now available in Cantonese.

VoiceMail Pro Enhancements

Unified Messaging Service (UMS)

VoiceMail Pro 4.2 replaces the existing Integrated Messaging Service (IMS) with a new Unified Messaging Service (UMS). This has two main uses:

- UMS is an Outlook integration solution that does not require the installation of client side components. VoiceMail Pro provides an IMAP Server to which IMAP Client capability of Microsoft Outlook can connect.

There is a new UMS User License. One valid license is required for each user of the service.

Check Digit Action

The current Check Digit Action has been enhanced to support the ability for the variables in the system to be validated against existing Extensions, Hunt Groups or Mailboxes. This is particularly important if constructing call flows that require unsupervised transfers to these targets, when there is a need to validate the existence of a target before committing a caller to that target.

Menu Action Enhancements

The current Menu action implementation requires additional call flow elements to be constructed to accommodate Timeout, invalid entries, and retry loops. VoiceMail Pro 4.2 has simple enhancements to the Menu Action providing an easier and quicker way of accommodating this capability.

Extended VBScript support

VBScript action will now support VB scripts of up to 10,000 characters.

Phone Manager Enhancements

Out-Calling control

In IP Office 4.2, Phone Manager Pro and Phone Manager PC Softphone provide the ability for users to change their Out-calling options, including the ability to set phone numbers and associated timeouts. The Out-calling information is passed between Phone Manager Pro 4.2 and VoiceMail Pro 4.2.

Voicemail Forward

Phone Manager Pro and PC Softphone allow users to forward their voicemail messages to other IP Office users.

Password Changing

Phone Manager Pro 4.2 can now interact with the IP Office to change the User password. The Phone Manager User password is also used by the TAPI driver, if installed. The Phone Manager Password mechanism also accommodates the requirements of the Avaya TAPI Driver when changed.

Softphone Vista Support

Phone Manager PC Softphone 4.2 supports operation on Windows Vista Business/Ultimate.

Softphone supports Login

Phone Manager PC Softphone 4.2 now allows users to login as an agent and therefore supports interaction with Compact Contact Center and Small Business Contact Center.

© 2008 AVAYA All rights reserved.

Four Sources of Music on Hold

IP Office 4.2 allows up to 3 alternate WAV file sources for music on hold, selectable for Incoming Call Routes or Hunt Groups (not available on the Small Office Edition). On IP500 systems, each source can be up to 90 seconds long (30 seconds on IP406 V2 and IP412).

Call Priority Enhancements

The Incoming Call Route setting "Priority" has been modified from a simple 1, 2, and 3 to now display the following making easier understanding of the priority value: -

- 1-Low
- 2-Medium
- 3-High

A new Call Priority Promotion Timer has been introduced to allow calls that have been queuing for longer than the timer setting to be promoted in priority. In addition, the Dial Extension short code has been enhanced to allow a transferred call to have its priority changed.

ETSI CHI

In IP Office 4.2, the Line SubType ETSI CHI is now available on all supported PRI cards. When this subtype is selected, channels can be allocated from 1-30 or 30-1 and further parameters may be configured per channel: Incoming/Outgoing Line Group ID, Channel Name, In/Out of Service, Direction (Incoming, Outgoing, Bothway), Bearer (Voice, Data, Any).

ETSI CHI stands for European Telecommunications Standards Institute - Call Handling Initiator.

Restrict Network Interconnect

When configured, this feature allows an IP Office to be programmed such that calls cannot be connected between private and public trunk interface groups. Used in some countries to comply with regulations.

T3 Telephone Enhancements

IP Office 4.2 supports the following enhancements for all variants of Avaya T3 telephones (T3 telephones are not supported on the Small Office Edition):

- Display Hunt Group membership status
- Edit phone directory entry
- Enhance T3 display “follow Me” target
- Visual Voice and Conferencing
- Program Function and Soft keys
- Night service button
- Voicemail and Voicemail on Toggle Button
- Line appearance keys
- Transfer to Voicemail
- Button Programming

Other Features

Telephone `Messages' Key Goes to Visual Voice

A new manager configuration option in IP Office 4.2 allows the messages key to initiate Visual Voice on supported phones (on non supported phones the previous behavior will continue). This configurable option will free the button programming required currently for Visual Voice.

SMDR Output from IP Office

IP Office 4.2 can now provide SMDR (Station Message Detail Recording) call logging information direct from the IP Office. As IP Office already has an IP based method for outputting Call Accounting records (CDR, or Call Detail Recording), this has been further refined to output SMDR.

The existing CDR tab has been renamed CDR/SMDR. SMDR cannot be output at the same time as CDR.

© 2008 AVAYA All rights reserved.

Reversal of Operation 'Outgoing Called Number Display'

IP Office 4.0 changed the feature phone display for outgoing calls from the user dialed digits to the digits sent to line. IP Office 4.2 reverses this change and the display for outbound external calls from the IP Office will be the user dialed digits. Where the call is made using the directory feature the number from the directory will be displayed as if the user dialed the digits.

Trunk Impedance Detection (North America only)

IP Office 4.2 includes a utility to test the impedance of analog trunks and propose appropriate gain settings. This could be useful if echo or noise problems are encountered on the line.

IP trunk fall back through ARS

A new Response Timeout option has been added to the IP Line, SIP Line and SES Line forms within IP Office 4.2 Manager to configure the timeout after which the line is considered unreachable. This enables fallback to an alternative trunk within Alternate Route Selection (ARS) within a reasonable time.

User Log Out Rule Change

In IP Office 4.2 facilitates the moving of users from one extension to another by allowing users to log out from their old extension, which will automatically result in their being logged in at the new extension (providing the new configuration has previously been merged).

IP Office XML Configuration Service

IP Office 4.2 introduces a backend configuration service component that runs on a Microsoft Server, to provide configuration services that can be accessed through a 3rd Party application via an XML interface. It enables:

- Secure configuration access to supported IP Office 4.2 platforms
- Retrieval of configuration data from the IP Office using secure TCP connection
- Exposure of configuration services via a web Service interface to obtain an XML representation of the configuration elements - this is limited to some User telephony settings, a subset of buttons and some Hunt Group settings.
- Transfer of the updated configuration back to the IP Office using a secure TCP connection

Support of this component is only available through the Avaya Developer Connection (DevConnect) program.

Support for Small Business Contact Center

IP Office 4.2 introduces several new parameters to support interaction with the Small Business Contact Center (SBCC), Avaya's forthcoming new contact center solution. The following parameters are visible within IP Office 4.2 Manager:

- "Customer Service Representative" checkbox in User/Telephony tab (and in User Rights)
- "CSR Hunt Group" checkbox in Hunt Group tab
- New System/SBCC tab with settings for Busy Not Available Reason Codes and After Call Work Time

New country support

New locales for German Swiss and Italian Swiss have been added, as has a Turkish locale.

Enhanced Outgoing Call Barring

Release 4.2 adds a password protected mechanism to allow a user to enable and disable the ability to make external outbound calls from an extension. This is supported through the introduction of a new Short Code and makes use of existing Login Codes.

© 2008 AVAYA All rights reserved.

Operating System Support

The following Microsoft Operating Systems and Applications will be supported by IP Office 4.2 core applications:

	IPO Manager	SSA	TAPI (1st Party)	TAPI (3rd Party)	TAPI WAV
Windows Server OS					
2000 Server (SP4)[1]	No	No	No	No	No
2003 Server (SP2)[2]	Yes	Yes	No	Yes	Yes
2008 Server 32-bit[3]	Yes	Yes	No	No	No
2008 Server 64-bit[3]	Yes	Yes	No	No	No
Windows Client OS					
2000 Professional (SP4)[1]	No	No	No	No	No
XP Professional (SP3)[4]	Yes	Yes	Yes	Yes	Yes
Vista Business/Enterprise (SP1)[5]	Yes	Yes	Yes	Yes	Yes
Vista Ultimate (SP1)[5]	Yes	Yes	Yes	Yes	Yes
Applications					
Outlook 2003	n/a	n/a	Yes	n/a	n/a
Outlook 2007	n/a	n/a	Yes	n/a	n/a

Notes:

1. Windows 2000 operating systems are no longer supported by IP Office. They may function but have not been tested with IP Office 4.2 and any faults reported with IP Office 4.2 will not be fixed.
2. 32-bit only. Small Business Server 2003 is supported for the same applications as Windows 2003 Server.
3. Subject to GA release of 2008 Server within acceptable timescales.
4. 32-bit only.
5. 32-bit only.

VoiceMail Pro 4.2 and all of its components will be able to run under the following operating systems (32-bit only. 64-bit versions of 2003/2008 Server and XP Professional are not supported.):

- Microsoft Windows XP Professional SP3
  – except VoiceMail Web Access
- Microsoft Vista Business / Ultimate SP1
  – except IMS, Campaigns and VoiceMail Web Access
- Microsoft Windows Server 2003 SP2
- Microsoft Windows Server 2008
  – except IMS

© 2008 AVAYA All rights reserved.

Notice about the support of Integrated Messaging (IMS):
IMS will not be supported under Vista and Server 2008 however it will be able to work together with Exchange Server 2007 but only on separate machines.

- Continued IMS support for IP Office releases 3.2 / 4.x
- Supported operating systems:
  - IMS Client: Windows XP SP3
  - IMS Server: Windows Server 2003 SP2
    - Windows Vista and Server 2008 will not be supported
- Support for:
  - Exchange Server 2000
  - Exchange Server 2003
  - Exchange Server 2007 (on a separate machine only, no installation on 64-bit O/S)

The following Microsoft Operating Systems and Applications will be supported by Phone Manager 4.2:

	Phone Manager
Windows Client OS	
2000 Professional (SP4)[1]	No
XP Professional (SP3)[2]	Yes
Vista Business/Enterprise (SP1)[3]	Yes
Vista Ultimate (SP1)[3]	Yes
Applications	
Outlook 2003	Yes
Outlook 2007	No

Notes:

1. Windows 2000 operating systems are no longer supported by IP Office across applications. They may function but have not been tested with IP Office 4.2 and any faults reported with IP Office 4.2 will not be fixed.
2. 32-bit only.
3. 32-bit only.

© 2008 AVAYA All rights reserved.

Voice Communication Solution Features

IP Office offers a comprehensive list of features and benefits for the small or mid-size business, including:

- **Full PBX features**
  Caller ID, Call Forwarding, Conference Calling, Voice Messaging and more.
- **Trunk Interfaces**
  A variety of network trunk interfaces, including E1, T1, PRI, ISDN, analog loop start and analog ground start for comprehensive network connectivity. Not all trunk types are available in all territories, please check for local availability.
- **Extensions**
  Support for a range of extensions, from 2 to 360 that provide sophisticated voice performance for new and growing businesses.
- **Telephones**
  A variety of telephones including analog, digital and IP hard and soft phones (wired and wireless) that provide the appropriate desktop or device phone for every need.
- **Advanced Call Routing**
  Incoming calls are directed to the best available person or messaging service, according to the company's unique criteria.
- **Alternate Call Routing**
  Ensures reliable handling of calls by selecting from analog, digital or VoIP trunks.
- **QSIG Networking**
  Standards-based multi-site networking to interoperate with other PABX's.
- **Integrated H.323 Gatekeeper and Gateway for converged communications**
  The IP Office acts as an IP telephony server with Quality of Service (QoS) support through DiffServ for routing and up to 128ms of Echo cancellation depending on Voice Compression Module (VCM) card fitted.
- **SIP Trunking**
  IP Office 4.0 and above supports SIP trunking to Internet Telephony Service Providers. This approach allows users with non-SIP phones to make and receive SIP calls.

© 2008 AVAYA All rights reserved.

Data Communication Solution Features

For offices with basic data networking needs, IP Office can provide a complete data communications and networking solution:

- **Internet Access**
  Firewall protected leased line or dial-up connectivity via PRI, T1 or WAN port: high-speed dialed access, direct leased line connections for high usage and Web site hosting, integral security, and efficient access to information and a larger business presence via the Web.
- **Routing**
  Integral Static or Dynamic (RIP I/II) routing for both Internet and Branch-to-Branch solutions.
- **Security**
  NAT (Network Address Translation) and built in firewall to protect your internal network and IPSec support allows secure VPN data transmission across public IP Networks using 3DES encryption.
- **DHCP**
  Automatic IP address allocation for local and remotely attached PC's and other devices, including IP phones.
- **Remote Access Server**
  Access to local LAN servers via optional two-channel V90 modem or digital trunks: individual firewall security, access control per user, and standards-based security enable remote workers.
- **LAN Switching**
  The IP500 and IP412 offer 2 switched Layer 3 Ethernet ports.
- **LDAP client support**
  For standards based directory synchronization with Phone Manager.

© 2008 AVAYA All rights reserved.

Applications Platform Features

IP Office provides big business benefits and enhanced productivity for small and mid-size businesses with a full complement of sophisticated applications. IP Office provides free-of-charge applications, including Phone Manager Lite and CTI interfaces. These free-of-charge applications can be upgraded to provide enhanced functionality through the purchase of license keys.

- **Operator SoftConsole**
  A graphical User Interface (GUI) for attendants on their PC desktop for call handling. Works with a telephone and is an easy way to learn and use sophisticated tools in a comfortable environment.
- **Phone Manager**
  A powerful desktop application for the IP Office, available in Lite, Pro, and PC Softphone versions to allow you to control and manage phone calls from your Windows desktop.
- **Open CTI interfaces**
  IP Office has a built in TAPI server that integrates easily with popular contact management applications such as Outlook, ACT!, GoldMine and Maximizer. Sophisticated custom applications can be rapidly developed and deployed with our full software development kit.
- **Voicemail**
  Callers can always be answered with a personal voicemail greeting before a message is taken and message notification set. Messages can be shared (forwarded) with colleagues and retrieved by any phone capable of tone dialing. When used with Phone Manager Pro, the PC can be used to control message playback.
- **Integrated Voicemail to Email Presentation**
  Voice messages can be copied into email messages and delivered into the email system. IP Office uses SMTP or MAPI to deliver a copy of the voice message.
- **Unified Messaging**
  The Unified Messaging Service within VoiceMail Pro provides a higher level of integration to synchronize both voicemail and email inboxes.
- **VoiceMail Pro Web Access**
  Together with UMS comes the ability to access voice mails via an internet browser. Into the browser voice mails can be managed: listen to them via the PC speaker or the desk phone, forward, delete, mark as read / unread are some of the possibilities the Voice Mail Web Access provides.
- **Auto-Attendant**
  Simplify service for administrators with this easy-to-use feature with the ability to construct customized automated services allowing callers to efficiently navigate the system, and reach the right person, without the assistance of an operator. Available with VoiceMail Pro and with Embedded Voicemail for IP500, IP406 V2 and Small Office Edition.
- **Interactive Voice Response (IVR) and Text to Speech**
  Create automated customized systems allowing callers to interact with business information, for example, reading email, account enquiry systems, automated ordering systems, ticket purchasing systems, PIN number checking, remote time sheet management, etc. Enhance theses systems by using Text To Speech to read information back to callers.
- **Queue Manager and Campaign Manager**
  Powerful voice and IVR applications for the Contact Center that facilitate agent and traffic management for better productivity and customer service.
- **Compact Business Center**
  Report on overall system performance and basic call center functionality for up to three workgroups with quality of service reports, selected group reports, simple installation, and more.
- **Compact Contact Center**
  This is the IP Office Contact Center option, with a full customer management toolset including real time agent, system, group management, standard and custom reporting. It provides real time tracking and analysis, options for agent connection, and remote agent support and wallboards for installations of up to 75 agents.

© 2008 AVAYA All rights reserved.

Management Tools

The IP Office solution (phone system, router/firewall/DHCP server) is easily managed through the IP Office Manager. IP Office Manager is a Windows PC software application that connects to the IP Office system using TCP/IP. It can be on the same LAN as the IP Office, remote on the WAN, or connected via the Remote Access Server with a Telephone Adaptor, Router or the optional internal modem.

The System Status application is a useful diagnostic tool that provides enhanced details about equipment and resources in the IP Office system. This includes indication of alarms and details of current calls in progress for local or remote diagnostics.

Scalable Platform

The "all-in-one" IP Office Family — servers, media modules, trunk interface cards and software applications — give small and mid-size businesses the options they want to meet today's communications needs and plans for the future.

- **Avaya IP Office 500**
  Modular, flexible chassis which supports up to 32 extensions (up to 272 with expansion modules), with capacity for up to 16 analog trunks or 8 digital trunks (up to 192 T1 channels or 240 E1 channels) using internal daughter cards. Up to 8 Expansion Modules may be added to provide a combination of up to 272 analog, digital or IP extensions, with additional analog trunks through external Analog 16 modules. Features include 128 optional voice compression channels, 2 independently switched LAN ports and an optional Embedded Messaging card.
- **Avaya IP Office IP412**
  Supports 12 Expansion Modules providing a combination of up to 360 analog, digital or IP extensions, with capacity for 8 analog trunks or 4 digital trunks (up to 96 T1 channels or 120 E1 channels). Additional analog trunks can be added using IP400 Analog 16 modules. Features include 60 optional voice compression channels, 2 independently switched LAN ports, and 108 data channels. An Internal Modem Card can be added to answer up to 12 V.90 analog modem calls.

© 2008 AVAYA All rights reserved.

Telephone Options

IP Office supports multiple telephone solutions, giving the small and mid-size business maximum flexibility to choose according to their current and future needs:

- **IP Telephones**
  IP Office's integral H.323 Server supports Avaya 1700 and 5600 Series IP telephones, selected Avaya 4600 series IP telephones, Avaya T3 series IP telephones, Avaya 3600 series Wireless VoIP telephones and Phone Manager PC Softphone.
- **Digital Telephones**
  IP500 Digital Station 8 cards plus IP Office Digital Station 16 or 30 Expansion Modules support the Avaya 5400 Series of digital phones and Avaya T3 Series telephones, plus selected 2400 and 6400 Series phones. The IP Office Digital Station expansion modules also support existing selected 2400, 4400, 6400 Series phones
- **Analog Phones**
  IP500 Phone 2 or 8 cards plus IP Office Phone 8, 16 or 30 Expansion Modules support standard analog phones, faxes and modems, with support for calling line identification and message waiting indication where these services are provided.
- **Wireless Telephones**
  Avaya IP DECT base stations can be added to support the Avaya IP DECT 3701 and 3711 telephones. The IP Office Digital Station interfaces support the Avaya 3810 telephone. The Avaya 3600 series wireless VoIP telephones are also supported.

Application and Feature Licensing

IP Office is an applications platform and includes a number of applications as part of the solution. These Lite versions of applications do not require any additional licensing, but upgrades to Pro versions or optional applications will need additional IP Office licenses to operate. The licensed applications require both a license key, a unique number that enables the application to run, and a feature key. The feature key is an electronic key installed on the IP Office system that determines which licensed applications can run.

Licensed applications are supplied in two forms; time limited trial licenses and full indefinite licenses. Trial licenses allow applications to run in fully functional form for 45 days (from the date of license generation), after which time they cannot be used until upgraded at cost to the full license but can be ordered at any time during the product ownership. Trial licenses are available for:

- Avaya Text To Speech (1 port).
- Centralized Voicemail with ACM.
- Conferencing Center.
- Mobility (5 user).
- Phone Manager PC Softphone (10 user).
- Phone Manager Pro (10 user).
- SoftConsole (1 user).
- Third Party Text To Speech (1 port).
- VB Scripting.
- VoiceMail Pro (4 ports)
- VoiceMail Pro Networked Messaging
- VoiceMail Pro Unified Messaging Service (10 user)
- VPN IPSec/L2TP.
- 3rd Party Database/IVR.
- SIP Trunking (1 trunk).
- Standard Edition upgrade to Professional Edition (for IP500 only).
- Voice Networking (4 channels, for IP500 only).
- Advanced Networking.
- VPN Phone (1 user).

Note: ContactStore has a 45 day trial built into the software and therefore does not require a separate license key, but this 45 day trial runs from when the software is installed.

© 2008 AVAYA All rights reserved.

Chapter 2.
IP Office Platform

© 2008 AVAYA All rights reserved.

2. IP Office Platform

IP Office is a modular communications solution that scales from 2 to 360 users. It provides a hybrid PBX with both Time Division Multiplexing (TDM) and IP phone support that can be used in either mode or both concurrently. IP Office has data capabilities built in, providing IP routing, switching and Firewall protection between LAN and WAN. IP Office has an integrated software applications suite that delivers a contact center, voice and email messaging, Interactive Voice Response, conferencing and computer telephony integration.

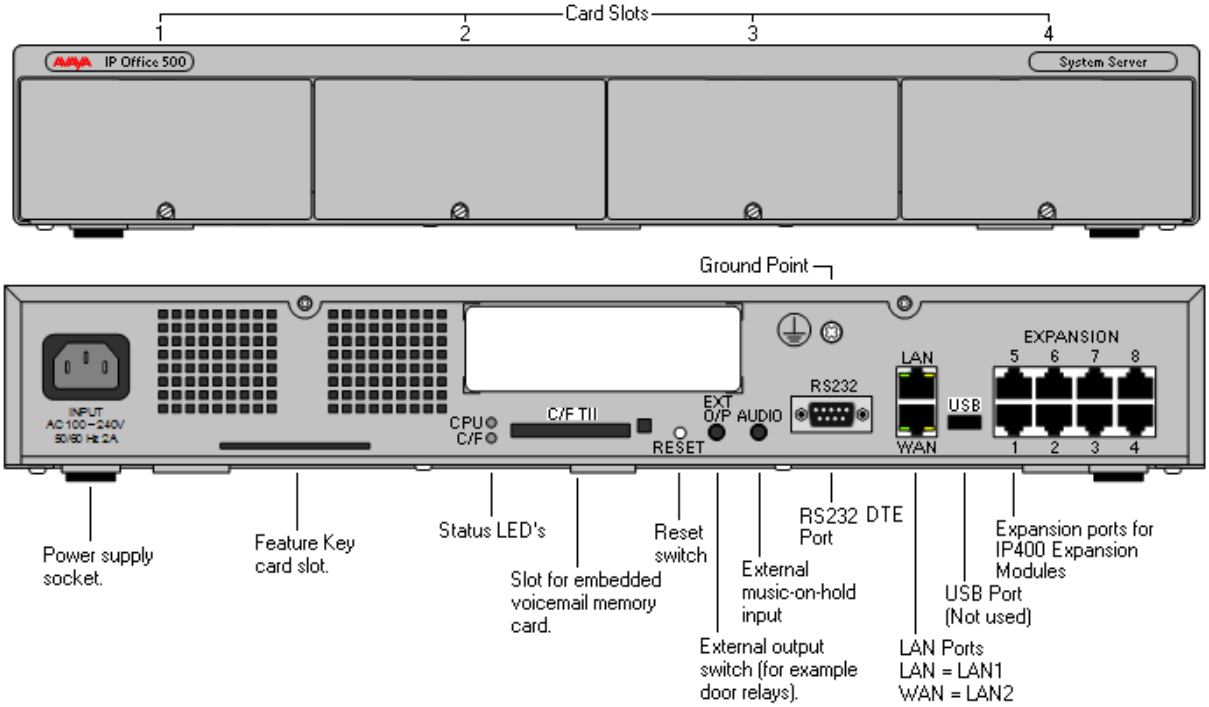
IP Office solutions are built from hardware units and application software. Hardware provides the connectivity for voice and data circuits and processor units for the solution software. Each IP Office solution will require a system control unit (IP500 or IP412), trunk connections to service provider, and expansion modules for TDM phone cabling. IP Phones connect over LAN connections to the IP Office solution.

© 2008 AVAYA All rights reserved.

IP Office 500 Control Unit

With a greater VCM channel capacity and performance, the IP Office 500 (IP500) is the most suitable of the IP Office range for IP Telephony applications. It also provides an entry level offer into the IP Office family through IP Office Standard Edition software. The IP500 also differs from the IP412 by providing a greater trunk expansion capability of up to eight four PRI interfaces (maximum 192/240 trunks). The IP500 is a stackable unit with an optional 19" rack mounting kit and an optional wall mounting kit for smaller configurations. The IP500 includes:

- 4 slots to house a mixture of extension cards and VCM cards
  - Digital Station 8 card
  - Phone 2 and Phone 8 cards
  - VCM-32 and VCM-64 cards
- Optional trunk daughter card support:
  - Analog Trunk Module 4 card
  - BRI-4 and BRI-8 cards (2 x 2B+D and 4 x 2B+D channels respectively)
  - Single and Dual Universal PRI cards
- Support for IP400 trunk and VCM cards using a Legacy Card Carrier
- Slot for smart card Feature Key – required for system operation as well as licensing of optional features.
- 9-pin DTE Port for maintenance.
- Support for up to 8 IP500 Expansion Modules (requires upgrade to Professional Edition):
  - Phone modules (8, 16, 30)
  - Digital Station modules (16, 30)
  - Analog Trunk Module 16
  - BRI So8 module
  - IP400 expansion modules (not WAN3 10/100 or Network Alchemy modules)
- External O/P socket supporting two relay on/off switch ports, e.g. for door entry systems.
- Audio input port for external music on hold source.
- 48 Data channels
- Up to 30 VoiceMail Pro ports
- Two 10/100 switched Ethernet ports (Layer 3).



© 2008 AVAYA All rights reserved.

IP500 Voice Networking License

QSIG, H.323 and SCN capabilities are not enabled by default in the IP500. An additional license is required to enable this functionality with 4 simultaneous networking channels (no channel limit for QSIG). Additional channels can then be licensed in increments of 4.

IP Office Standard Edition

By default the IP500 control unit runs a subset of full IP Office functionality called IP Office Standard Edition. In this mode the IP500 is restricted to a maximum of 32 users in the base control unit with no expansion. Supported options include Embedded Voicemail, Phone Manager Lite/Pro/PC Softphone, SoftConsole, TAPI, SMDR, SIP trunking, mobile twinning, VPN Phone and IP DECT, as well as licenses for voice networking (H.323 or SCN). IP Office Standard Edition does not support advanced applications (VoiceMail Pro, CCC, Conference Center, etc). This restriction can be removed by adding an IP Office Professional Edition Upgrade license to the configuration.

IP Office Professional Edition

By purchasing the upgrade license from Standard Edition to Professional Edition, additional functionality is enabled. This includes the ability to expand the system using up to eight external Expansion Modules, allowing the IP500 to support a maximum of 272 phones through a mixture of analog, digital or IP handsets. If additional analog trunks are required, these can be aggregated in groups of 16 on each analog expansion module. Note that the Professional Edition also enables the licensing of advanced applications such as VoiceMail Pro.
The following table shows which features are supported by Standard Edition and which require the upgrade to Professional Edition.

	Standard Edition		Professional Edition	
	Included	Optional	Included	Optional
Up to 32 users.	✓		✓	
IP phone support	✓		✓	
Phone Manager Lite	✓		✓	
TAPI	✓		✓	
SMDR	✓		✓	
64-way basic conferencing	✓		✓	
IP Office Manager	✓		✓	
System Status Application	✓		✓	
Voice Networking		✓		✓
Advanced Networking		✓		✓
Embedded Voicemail		✓		✓
Phone Manager Pro / PC Softphone		✓		✓
SoftConsole		✓		✓
SIP Trunking		✓		✓
IP DECT		✓		✓
Mobile Twinning and Mobility		✓		✓
VPN phone support		✓		✓
CTI Link Pro		✓		✓
32 - 272 Users supported.			✓	
Expansion Module support.			✓	
64-way Meet-Me Conferencing			✓	
VoiceMail Lite			✓	
VoiceMail Pro				✓
UMS				✓
ContactStore				✓
Conferencing Center				✓
CCC				✓
TAPI WAV				✓

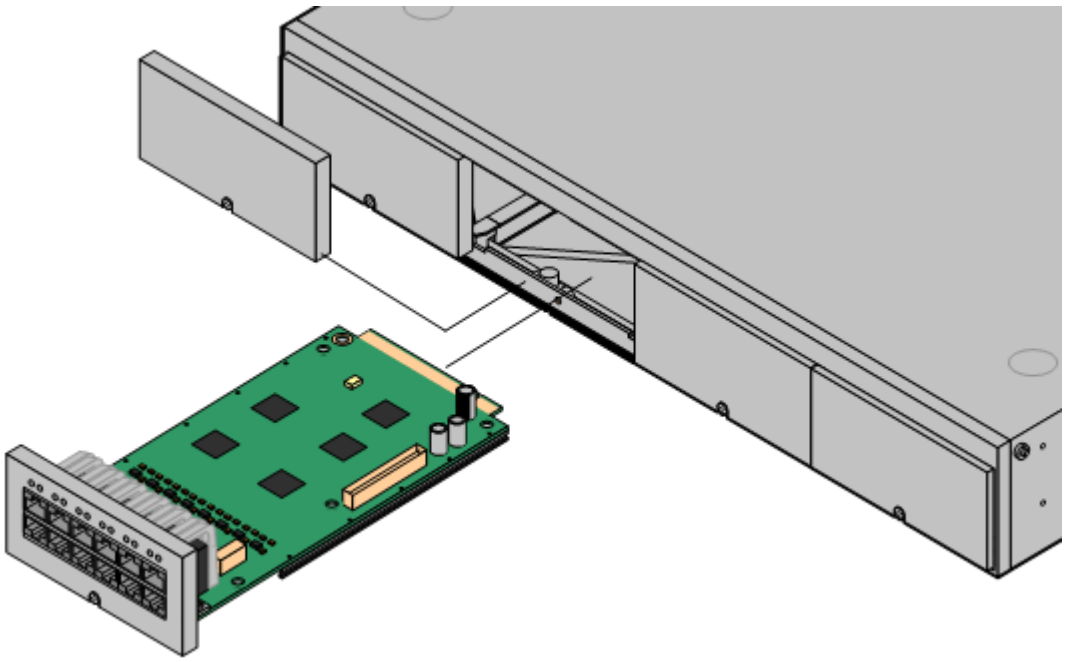
© 2008 AVAYA All rights reserved.

IP500 Cards

The IP500 control unit has 4 slots for the insertion of cards. These cards can be divided into two types; base cards and daughter cards. Base cards include a front panel and ports for cable connections. Daughter cards can be added to a base card in order to provide additional facilities (typically trunk connections).

The following base cards are available:

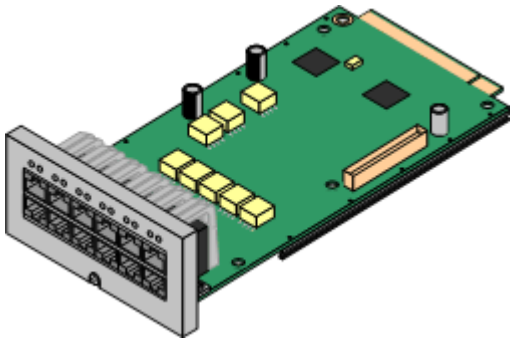
- **IP500 Digital Station 8 Card** 28 ***(Maximum 3).***
- **IP500 Analog Phone 2 Card and Phone 8 Card** 28 ***(Maximum 4).***
- **IP500 VCM Card** 29 ***(Maximum 2).***
- **IP500 Legacy Card Carrier** 29 ***(Maximum 2)*** 29.



© 2008 AVAYA All rights reserved.

IP500 Digital Station 8 Card

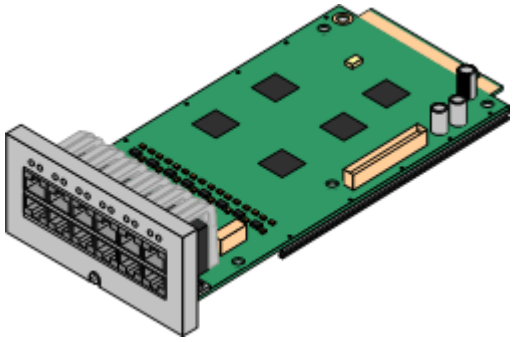
This card provides 12 RJ45 ports. The first 8 ports are DS ports for the connection of Avaya digital phones other than IP phones. The card can be fitted with an IP500 daughter trunk card, which then uses the additional 4 RJ45 ports for trunk connections.



- This card accepts one IP500 trunk daughter card of any type.
- This card supports selected Avaya 2400, 5400 and 6400 series phones, plus T3 phones(EMEA only) and 3810 wireless phones (North America only).
- 4400 Series phones (4406D, 4412D and 4424D) are not supported on this card, only on Digital Station expansion modules. Therefore a maximum of 240 x 4400 Series phones are supported in the system.

IP500 Analog Phone 2 Card

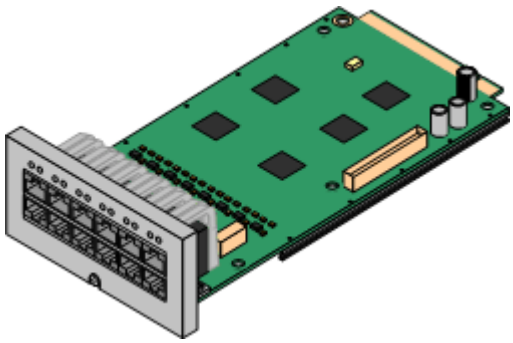
This card provides 2 analog extension ports (1-2) for the connection of analog phones. The card can be fitted with an IP500 daughter trunk card, which then uses the last 4 RJ45 ports (9-12) for trunk connections.



- This card accepts one IP500 trunk daughter card of any type.

IP500 Analog Phone 8 Card

This card provides 8 analog extension ports for the connection of analog phones. The card can be fitted with an IP500 daughter trunk card, which then uses the additional 4 RJ45 ports for trunk connections.

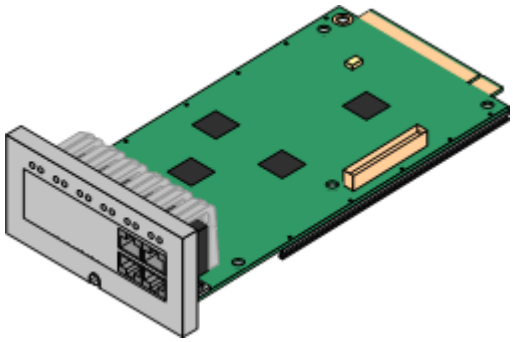


- This card accepts one IP500 trunk daughter card of any type.

© 2008 AVAYA All rights reserved.

IP500 VCM Card

This card provides voice compression channels for use with VoIP calls, SIP trunks and IP-based voice networking. The module is available in variants supporting up to 32 or 64 channels. The actual number of channels provided is controlled by the VCM Channels licenses entered into the IP500 system configuration.



Each VCM card has 4 VCM channels enabled by default. Further channels can be enabled up to the maximum (32 or 64) through adding one or more licenses (available in 4, 8, 16, 28 and 60 channel increments).

The maximum number of voice compression channels supported, using IP500 VCM base cards and / or IP400 VCM cards on IP500 Legacy Card Carriers, is 128.

Please note, the maximum number of channels which may be used on each IP500 VCM card card varies according to the codec used, as shown in the following table.

Maximum VCM channels available, by codec type:

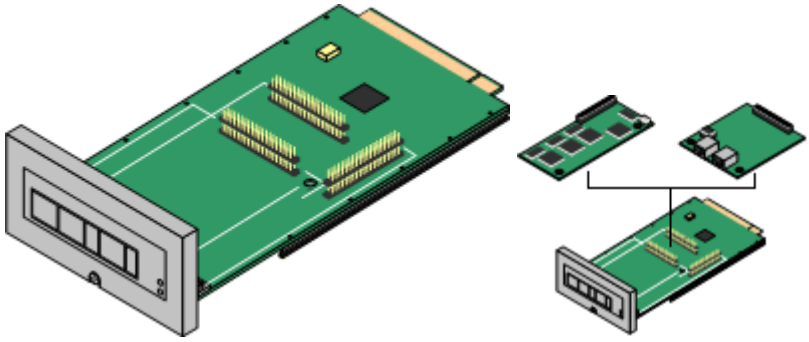
	VCM 32	VCM 64
G.711	32	64
G.729	30	60
G.723	22	44

The card can be fitted with an IP500 daughter trunk card, which uses the 4 RJ45 ports for trunk connections.

- This card accepts one IP500 trunk daughter card of any type.

IP500 Legacy Card Carrier

This card allows a variety of IP400 trunk and VCM cards to be used with the IP500 control unit. The front of the card includes a number of panels that can be snapped off to match the ports available when an IP400 trunk card is fitted.

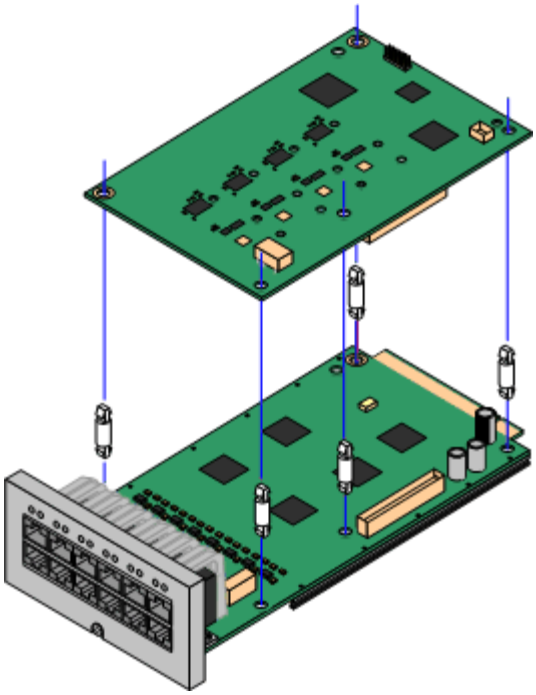


- This card does not accept any IP500 daughter trunk cards.
- The IP500 control unit can accept up to 2 IP400 trunk or VCM cards by mounting each card on an IP500 Legacy Card Carrier
- This card supports the following IP400 cards:
  - ✓ PRI T1
  - ✓ Dual PRI T1
  - ✓ PRI 30 E1 (1.4)
  - ✓ Dual PRI E1
  - ✓ PRI 30 E1R2 RJ45
  - ✓ Dual PRI E1R2 RJ45
  - ✓ ANLG 4 Uni
  - ✓ BRI-8 (UNI)
  - ✓ VCM 4
  - ✓ VCM 8
  - ✓ VCM 16
  - ✓ VCM 24
  - ✓ VCM 30

© 2008 AVAYA All rights reserved.

IP500 Trunk Cards

IP500 daughter trunk cards can be fitted to existing IP500 base cards to provide support for trunk ports. The daughter card uses the ports provided on the base card for cable connection. The addition of an IP500 daughter trunk card is supported on IP500 Digital Station, IP500 Analog Phone and IP500 VCM base cards. They are not supported on the IP500 Legacy Card Carrier base card.



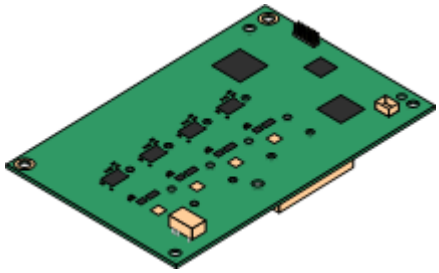
For those base cards that support daughter cards, there are no restrictions on the combination of card types. However in systems with both Analog Phone 8 base cards and analog trunk daughter cards, combining the two types are recommended as it then provides analog power failure support for one trunk/extension (Not applicable to the Analog Phone 2 base card).

Each daughter card is supplied with the spacer pegs required for installation and a label to identify the card's presence on the physical unit once installed.

- **IP500 Analog Trunk Card (Maximum 4).**
- **IP500 BRI Trunk Card (Maximum 4).**
- **IP500 Universal PRI Trunk Card (Maximum 4).**

IP500 Analog Trunk Card

This card can be added to an IP500 Digital Station card, IP500 Analog Phone base card, or IP500 VCM card. It allows that card to then also support 4 analog loop-start trunks. It also provides one analog V.32 modem.

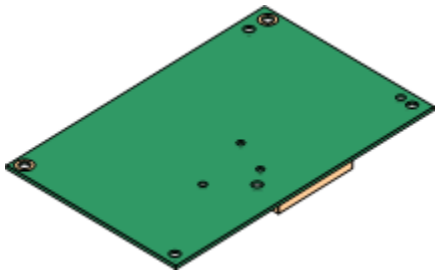


- When fitted to an IP500 Analog Phone 8 base card, the combination supports 1 power failure extension to trunk connection.

© 2008 AVAYA All rights reserved.

IP500 BRI Trunk Card (Euro ISDN)

This type of card can be added to an IP500 Digital Station card, IP500 Analog Phone card, or IP500 VCM card. It allows that card to then also support up to 4 BRI trunk connections, each trunk providing 2B+D digital channels. The card is available in 2 port (4 channels) and 4 port (8 channels) variants.

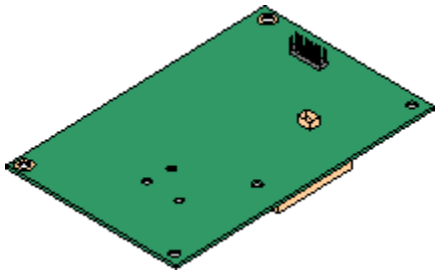


From IP Office release 4.2 onwards, the ports on IP500 BRI cards can be individually configured to operate as an S-interface (S0 mode). When set as an S interface the settings for the Line will be exactly the same as for a circuit on an So8 Expansion Module.

These trunks support the mobility features of Mobile Call Control and one-X Mobile client support introduced in Release 4.2.

IP500 Universal PRI Trunk Card

This type of card can be added to an IP500 Digital Station card, IP500 Analog Phone card, or IP500 VCM card. It allows that card to then also support primary rate digital trunk connections. Available in single and dual versions the IP400 PRI card provides single and dual primary rate trunk interfaces respectively. The PRI is configurable for T1, E1 or E1R2 MFC use depending on the territory.



Details of the supported ISDN supplementary services and protocols for each PRI are given in the 'Public and Private Voice Networks' section.

The IP500 Universal PRI trunk cards incorporate an integrated CSU/DSU. The CSU function allows the trunk to be put in loop-back mode for testing purposes. This can be set manually, using the monitor application, or automatically from a Central Office sending a Line Loop Back (LLB) pattern. The DSU function allows the T1 trunk to be shared between data and voice services.

Here is a summary of the capabilities of the card:

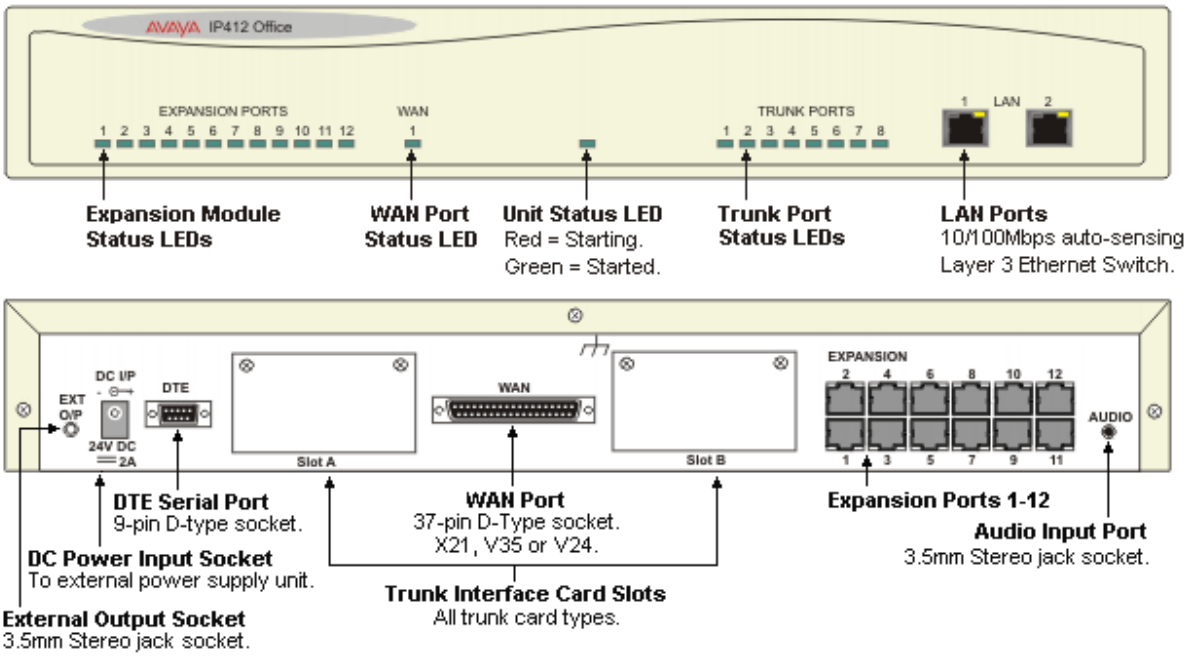
- Each card is configurable to connect to T1, E1 or E1R2 lines.
- The card is available in either a single or dual PRI variant. The single variant can support up to 24 T1 channels or up to 30 E1 channels. The dual variant can support up to 48 T1 channels or 60 E1 channels.
- On each card, 8 channels per interface are enabled by default. This means that the single PRI has 8 channels enabled, while the dual PRI has 16 channels enabled. Further channels may be enabled by the purchase of additional licenses in 2-channel or 8-channel increments.
- The IP500 PRI daughter card works on any IP500 VCM or extension base card (not the Legacy Card Carrier).
- Up to four Universal PRI cards can be installed in any combination in the IP500 chassis.
- Diagnostics capabilities:
  - Visual indicators to show service state
  - Physical test points to monitor traffic.
- These trunks support the mobility features of Mobile Call Control and one-X Mobile client support introduced in Release 4.2.

© 2008 AVAYA All rights reserved.

IP Office IP412 Control Unit

With a greater internal data transfer capability than the IP406 V2, the IP412 is more suitable for meeting the needs of the small contact center or businesses with a CRM focus. The IP412 differs from the IP4060 V2 by providing a greater trunk expansion capability of up to four PRI trunks. The IP412 is a stackable unit with an optional 19" rack mounting kit. The IP412 includes:

- 9-pin DTE Port (for maintenance or Feature Key connection for application licensing).
- X.21/V35 WAN interface.
- Support for up to 12 IP Office Expansion Modules:
  - Phone modules (8, 16, 30)
  - Digital Station modules (16, 30)
  - Analog Trunk Module 16
  - So8 module
- External O/P socket supporting two relay on/off switch ports, e.g. for door entry systems.
- Audio input port for external music on hold source.
- Two trunk interface card slots for analog, BRI, PRI (T1, E1) or CAS (E1R2)
- 2 internal sockets for IP Telephony expansion – voice compression modules (from 4 to 30 channels)
- Internal socket for internal modem (2 or 12) for Remote Access Services
- 108 Data channels
- Up to 30 VoiceMail Pro ports
- Two 10/100 switched Ethernet ports (Layer 3).



© 2008 AVAYA All rights reserved.

Expansion Modules

Through support of up to twelve external Expansion Modules, IP412 can be enhanced to support a mixture of analog, digital or IP phones, to maximum of 360 phones in any combination.

If additional analog trunks are required, these can be aggregated in groups of 16 on each analog expansion module.

Data Channels

A Data Channel is used for Remote Access (RAS), Internet Access, and Voicemail sessions. A data channel is an internal signaling resource used whenever a call is made from the IP network to an exchange line (Central Office). For example, four people surfing the Internet will use a single data channel since they all share the same line to the ISP. Two people remotely accessing the Office LAN from home will use two data channels since they have dialed in on separate lines. IP extensions do not use data channels. Data channels are used for voicemail connections with a maximum of 30 available for VoiceMail Pro on a IP412.

Modems and Voice Compression modules

You can add additional hardware to the IP412 system to add one modem card (12 V.90 modems) and 2 Voice Compression Modules (VCM). Each VCM supports from 4 to 30 simultaneous Voice over IP sessions and is used for either providing networking between sites over a Wide Area Network or supporting IP Telephones and Soft phones.

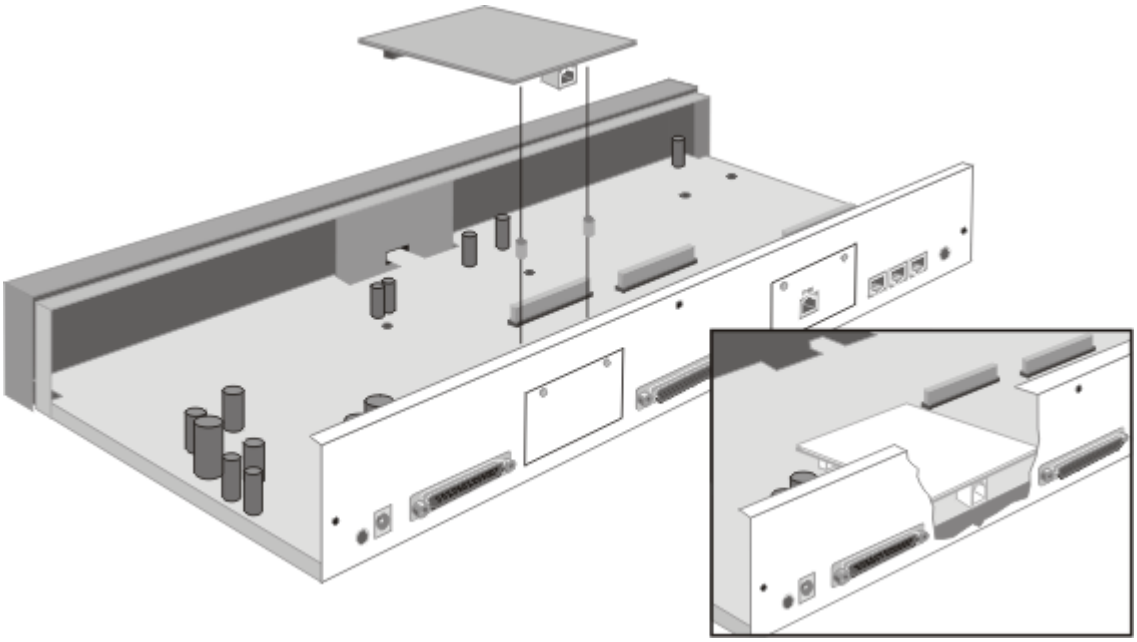
© 2008 AVAYA All rights reserved.

IP400 Trunk Interface Cards

IP400 trunk interface cards fit into the card slots on the Small Office Edition, IP406 V2 and IP412 control units and in any slot of the IP500 when combined with the IP500 Legacy Card Carrier. They provide analog, ISDN or CAS trunk connectivity. Not all interfaces are available in all territories. The following table shows how many of each card type are supported by each of control unit.

	Small Office Edition	IP406 V2	IP412	IP500*
IP400 Universal Analog Trunk 4	✗	2	2	2
IP400 Quad BRI	1	2	2	2
IP400 PRI E1	✗	2	2	2
IP400 Dual PRI E1	✗	1 (Slot A)	2	2
IP400 E1R2	✗	2	2	2
IP400 Dual E1R2	✗	1 (Slot A)	2	2
IP400 PRI T1	1	2	2	2
IP400 Dual PRI T1	✗	1 (Slot A)	2	2

* Each card requires the use of an IP500 Legacy Card Carrier. IP500 does not support E1R2 coax cards.



IP400 BRI Card

The BRI trunk card provides 4 Basic Rate ISDN T interfaces (8 trunks). Details of the supported ISDN supplementary services on BRI interfaces are given in the 'Public and Private Voice Networks' section.

© 2008 AVAYA All rights reserved.

IP400 PRI Cards (T1/E1/E1R2)

Available in single and dual versions the IP400 PRI card provides single and dual primary rate trunk interfaces respectively. The PRI is available in T1, E1 and E1R2 MFC variants depending on the territory. The dual version is only supported on the IP412, in slot A of the IP406 V2 and in the IP500 using an IP500 Legacy Card Carrier. The Small Office Edition only supports the single T1 PRI card (not E1 or E1R2).

Details of the supported ISDN supplementary services and protocols for each PRI are given in the 'Public and Private Voice Networks' section.

T1 trunk cards incorporate an integrated CSU/DSU. The CSU function allows the trunk to be put in loop-back mode for testing purposes. This can be set manually, using the monitor application, or automatically from a Central Office sending a Line Loop Back (LLB) pattern. The DSU function allows the T1 trunk to be shared between data and voice services.

IP400 Universal Quad Analog Trunk (LS) Card

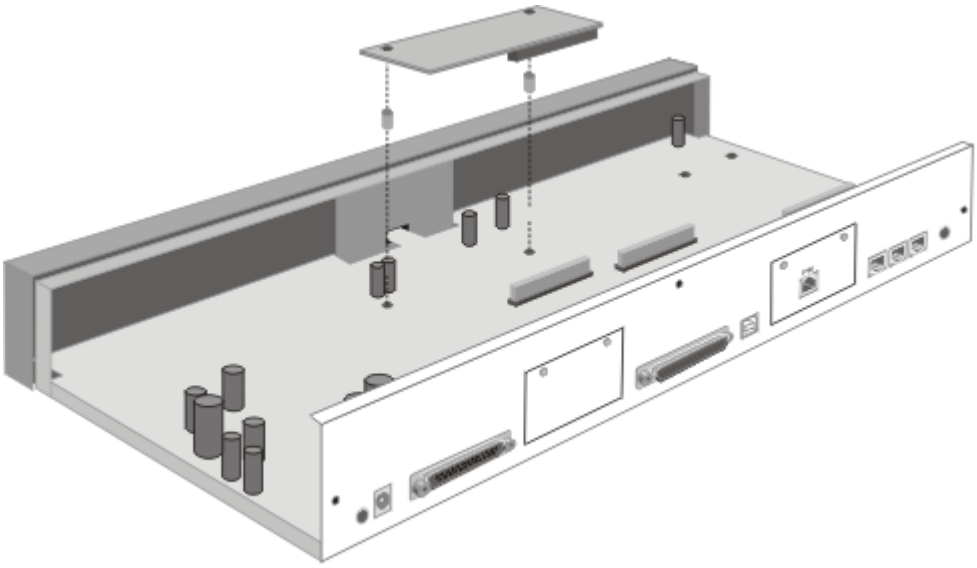
This card provides four analog trunk ports. These are 2-wire loop start interfaces and are available in all territories. This card supports Caller ID where provided. With IP Office R3.1 and later, this module supports optional 16ms echo cancellation.

Please note that ground start analog trunks are supported via the IP Office Analog Trunk 16 Expansion Module.

© 2008 AVAYA All rights reserved.

Internal Daughter Cards

Internal Daughter Cards are fitted inside the IP406 V2 and IP412 control units.



IP400 Voice Compression Module – 4/8/16/24/30 ports

The Voice Compression Module (VCM) is used for Voice over IP (VoIP) applications in the IP406 and IP412 control units. Five VCM variants are available supporting 4, 8, 16, 24 and 30 channels of compression. The echo cancellation capabilities of the VCM cards vary. The VCM 4, 8, 16 and 24 cards offer 64ms of echo cancellation while the VCM 30 card offers 25ms.

On IP Office - Small Office Edition systems, either 3 or 16 VCM channels are built-in and cannot be changed. The IP406 V2 supports a single VCM while the IP412 can have any two VCMs installed.

The following table shows how many of each card are supported by each platform.

	IP406 V2	IP412	IP500
IP400 VCM 4	1	2	2*
IP400 VCM 8	1	2	2*
IP400 VCM 16	1	2	2*
IP400 VCM 24	1	2	2*
IP400 VCM 30	1	2	2*
IP400 VCM 5/10/20 ***(no longer sold)***	1	2	✗
IP500 VCM 32	✗	✗	2
IP500 VCM 64	✗	✗	2

*Each card requires the use of an IP500 Legacy Card Carrier.

IP400 Internal Modem Card

An internal modem card with 12 modems can be installed in both the IP406 V2 and IP412 to provide dial-up capacity that is better matched to remote access requirements of customers. The Internal Modem card allows up to 12 simultaneous V.90 (56kbps) analog modem calls into the IP Office.

© 2008 AVAYA All rights reserved.

External Expansion Modules

Unless otherwise stated, each of these modules may be used with the IP500, IP406 V2 and IP412.

- **IP500 Phone Expansion Module** 38
  Available in two variants for 16 or 30 analog extensions with calling line presentation.
- **IP500 Digital Station Expansion Module** 40
  Available in two variants for 16 or 30 digital extensions for Avaya series digital telephones.
- **IP500 BRI So8 Expansion Module** 39
  Available regionally, offering 8 BRI S-interfaces for ISDN connection.
- **IP500 Analog Trunk 16 Expansion Module** 39 ***(US version only)***
  Provides 16 analog loop start or ground start trunks, with power failover of two trunks.
- **IP400 Phone Expansion Module** 41
  Available in three variants for 8, 16 or 30 analog extensions with calling line presentation.
- **IP400 Digital Station Expansion Module** 42
  Available in two variants for 16 or 30 digital extensions for Avaya series digital telephones.
- **IP400 So8 Expansion Module** 302 *(No longer sold)*
  Available regionally, offering 8 BRI S-interfaces for ISDN connection.
- **IP400 Analog Trunk 16 Expansion Module** 43
  Available in one variant for 16 analog loop start or ground start trunks, with power failover of two trunks.
- **IP400 WAN3 10/100 Expansion Module** 302 *(No longer sold)*
  Provides 3 wide area interfaces and connects to IP Office via Ethernet. A maximum of 2 WAN3 10/100 modules are supported on the IP406 V2 and IP412. It is not supported on the IP500 or Small Office Edition.

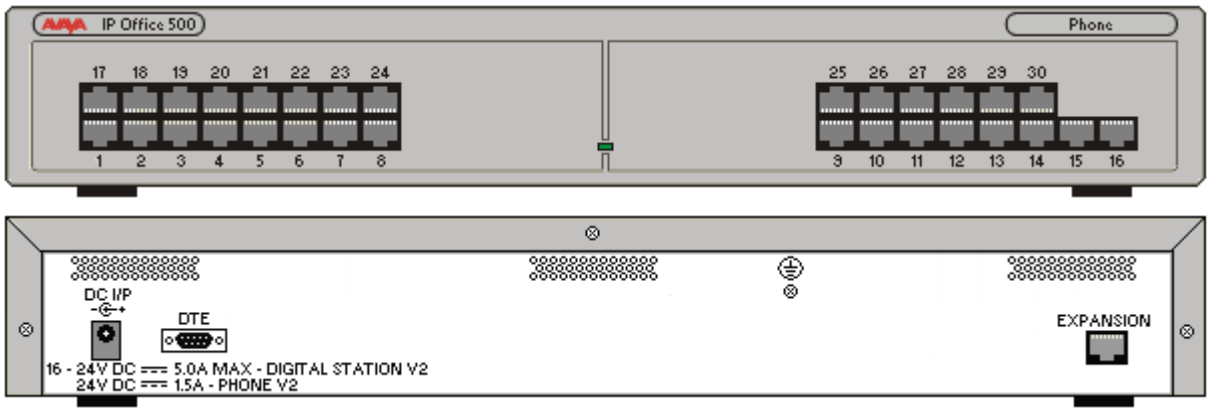
© 2008 AVAYA All rights reserved.

IP500 Phone Expansion Module

This module provides additional analog telephone interfaces:

- Two Wire.
- DTMF signaling (No rotary or Loop Disconnect).
- Time Break Recall (No Earth Loop Recall).
- Caller ID capable.
- Message Waiting Indication (MWI) capable - High Voltage, Pulsed High Voltage, Line Reversal.

The IP500 Phone module is available in 2 versions, giving 16 or 30 extensions. Telephones can be located up to 1km from the control unit. For extensions located "out-of-the-building" additional line protection will be needed. For more information on cabling and out of building guidelines, see the IP Office Installation Manual.



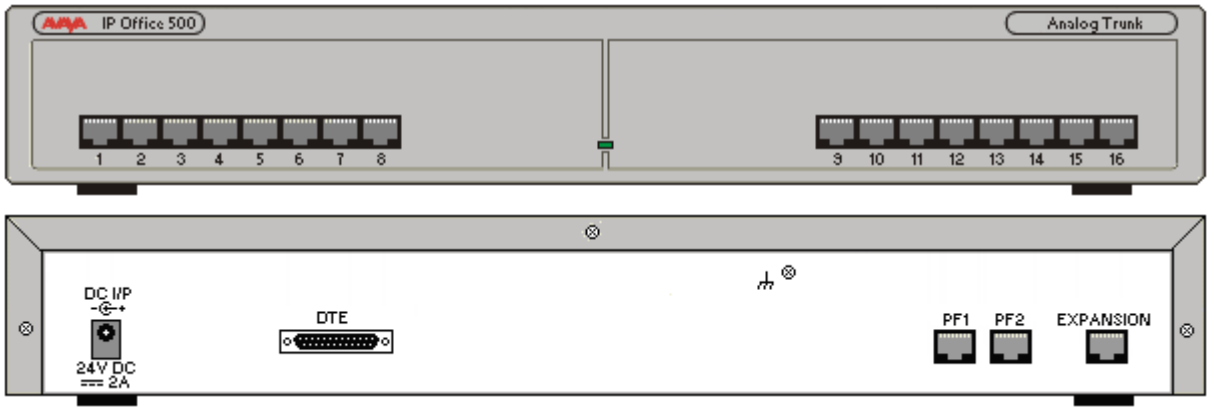
- IP Office Phone Modules provide support for a variety of analog MWI methods. These methods are 51V Stepped, 81V, 101V and Line Reversal. The 101V method is only supported when using a Phone V2 expansion module.
- Each analog port can support a device of maximum 1 REN.
- On analog ports, call information is sent while the phone is ringing, and cannot be updated during a call or set on an outbound call (the phone may do a local match but this is not controlled by the IP Office). The primary purpose of displays is to give information about incoming calls. Where the Caller Display standard chosen supports the delivery of text (extension name) as well as the number, both are delivered.
- An analog extension port can be set for external Paging operation. It does not operate like a normal extension and is connected to external equipment through an isolation device. The Port will always be busy so it cannot be called directly and can only be accessed by using a short code. When not receiving a Page the port will remain silent, when being paged the page tone is sent before the speech path is opened.

© 2008 AVAYA All rights reserved.

IP500 Analog Trunk 16 Module

This expansion module provides an additional sixteen Loop Start or Ground Start two-wire analog trunks. (Ground start trunks are not available in all territories) The first two trunks on the module which are automatically switched to power fail sockets on the rear of the unit in the event of power being interrupted must be loop start for correct power fail operation.

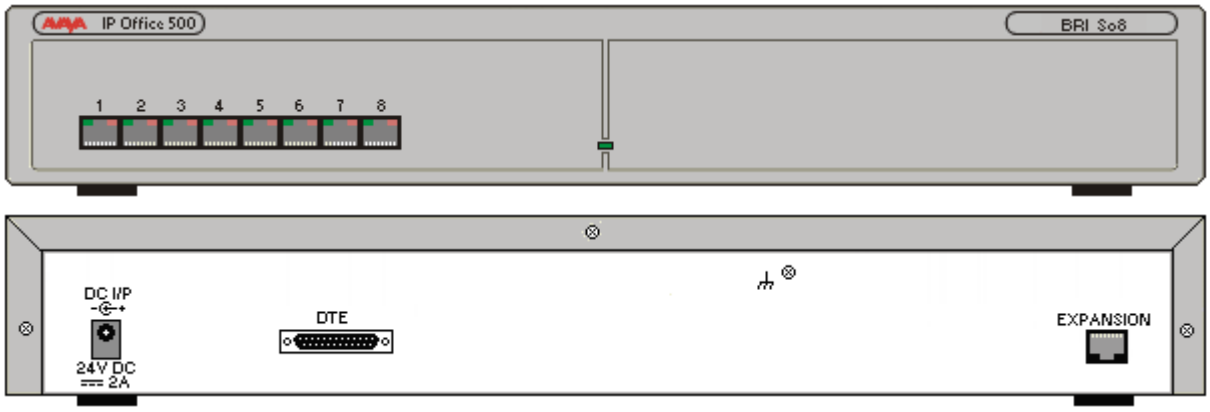
For installations in a rack, this module requires the IP500 Rack Mounting Kit. The IP500 Analog Trunk 16 Module is functionally identical to the IP400 Analog Trunk 16 Module.



IP500 BRI So8 Module

The IP500 BRI So8 module provides 8 S-Bus interfaces for Basic Rate ISDN devices, such as video conferencing, fax servers or ISDN telephones.

For installations in a rack, this module requires the IP500 Rack Mounting Kit. The IP500 BRI So8 Module is functionally identical to the IP400 So8 Module.



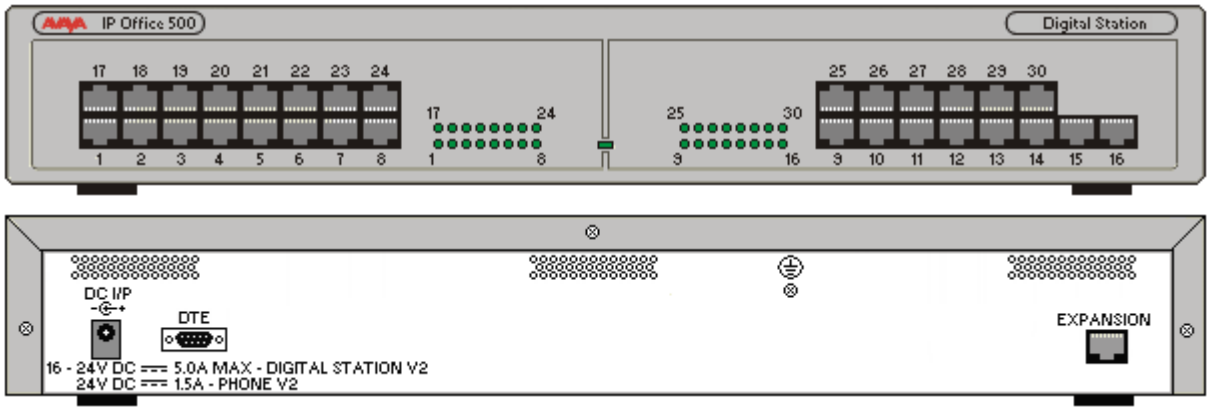
The IP500 BRI So8 expansion module supports both point-to-point and point-to-multipoint connections. A maximum of 10 terminal endpoints identifiers (TEIs) are supported on each bus.

© 2008 AVAYA All rights reserved.

IP500 Digital Station Module

This expansion module provides additional Digital Station (DS) ports for selected Avaya 2400, 4400, 5400, 6400, T3 (EMEA only) series phones and 3810 wireless phones (North America only). The IP500 Digital Station module is available in 2 variants; 16 or 30 extensions.

For installations in a rack, this module requires the IP500 Rack Mounting Kit. The IP500 Digital Station Module is functionally identical to the IP400 Digital Station V2 Module.



- Telephones can be located up to 1km from the control unit. For extensions located "out-of-the-building" additional line protection will be needed. For more information on cabling and out of building guidelines, see the IP Office Installation Manual.
- For systems where Direct Station Select (DSS) Units are being used, IP Office supports a maximum of:
  - Eight EU24 and or EU24BL per system.
  - Two XM24 units on each Digital Station expansion module to a maximum of 10 XM24 units per system.
  - Two 4450 units on each Digital Station expansion module to a maximum of 10 4450 units per system.
  - T3 DSS units.

See the Telephones Section 46 for specific limits on the number of each type of telephone supported on DS modules.

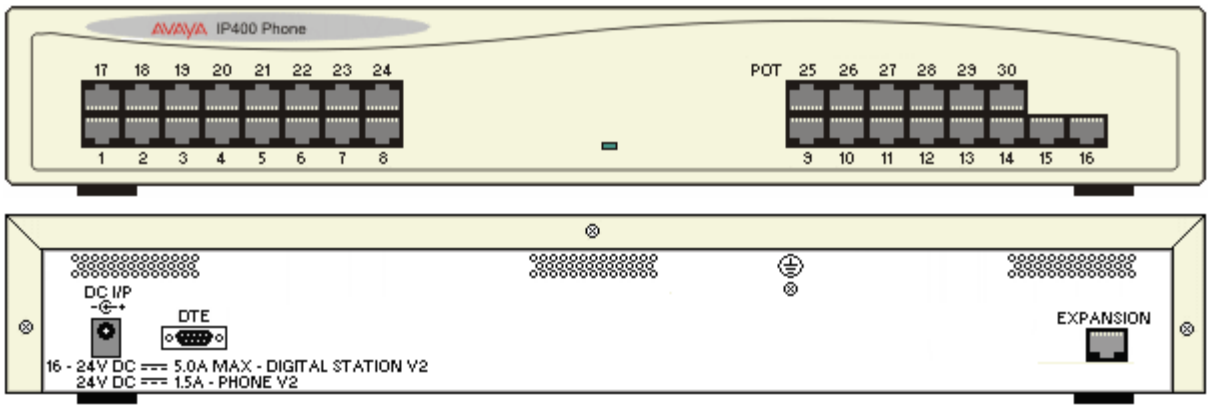
© 2008 AVAYA All rights reserved.

IP400 Phone Module

This module provides additional analog telephone interfaces:

- Two Wire
- DTMF signaling (No rotary or Loop Disconnect)
- Time Break Recall (No Earth Loop Recall)
- Caller ID capable
- Message Waiting Indication (MWI) capable - High Voltage, Pulsed High Voltage, Line Reversal

The IP400 Phone module is available in 3 versions, giving 8, 16 or 30 extensions. Telephones can be located up to 1km from the control unit. For extensions located "out-of-the-building" additional line protection will be needed. For more information on cabling and out of building guidelines, see the IP Office Installation Manual.

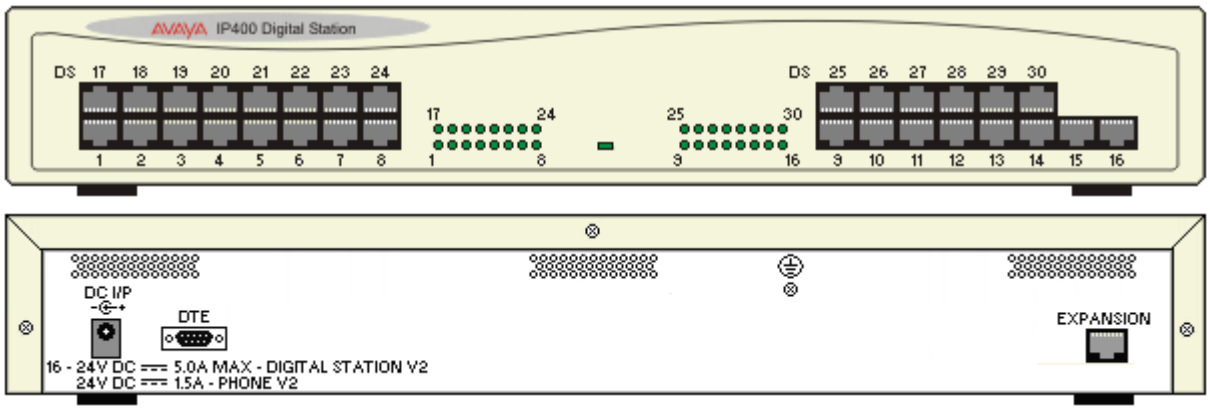


- IP Office Phone Modules provide support for a variety of analog MWI methods. These methods are 51V Stepped, 81V, 101V and Line Reversal. The 101V method is only supported when using a Phone V2 expansion module.
- Each analog port can support a device of maximum 1 REN.
- On analog ports, call information is sent while the phone is ringing, and cannot be updated during a call or set on an outbound call (the phone may do a local match but this is not controlled by the IP Office). The primary purpose of displays is to give information about incoming calls. Where the Caller Display standard chosen supports the delivery of text (extension name) as well as the number, both are delivered.
- An analog extension port can be set for external Paging operation. It does not operate like a normal extension and is connected to external equipment through an isolation device. The Port will always be busy so it cannot be called directly and can only be accessed by using a shortcode. When not receiving a Page the port will remain silent, when being paged the page tone is sent before the speech path is opened.

© 2008 AVAYA All rights reserved.

IP400 Digital Station V2 Module

This expansion module provides additional Digital Station (DS) ports for selected Avaya 2400, 4400, 5400, 6400, T3 (EMEA only) series phones and 3810 wireless phones (NA only). The IP400 Digital Station module is available in 2 variants; 16 or 30 extensions.



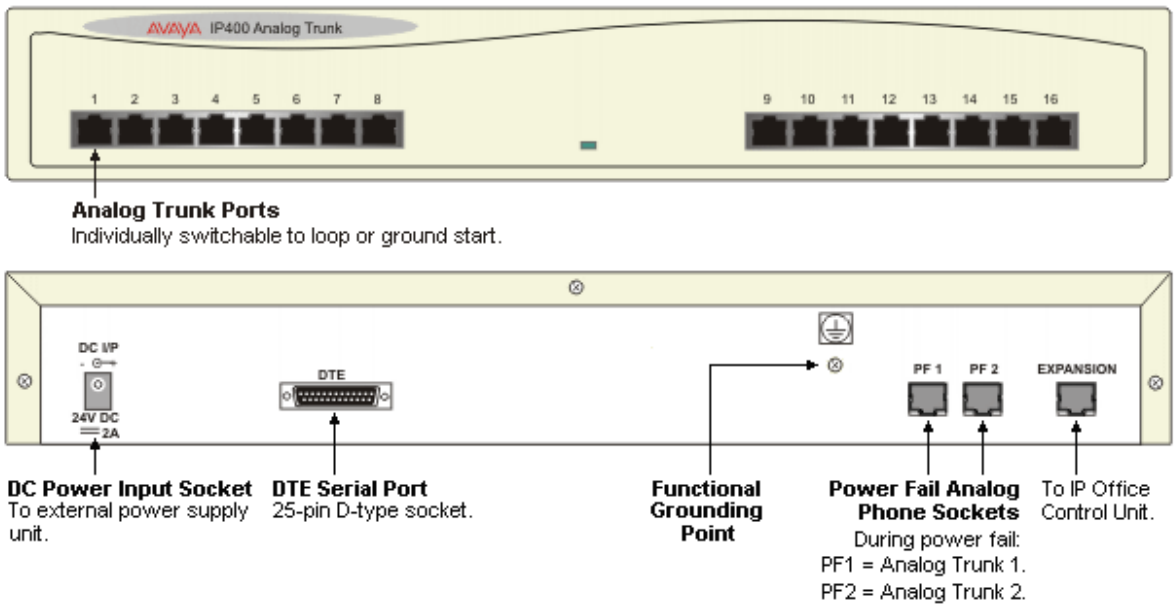
- Telephones can be located up to 1km from the control unit. For extensions located "out-of-the-building" additional line protection will be needed. For more information on cabling and out of building guidelines, see the IP Office Installation Manual.
- For systems where Direct Station Select (DSS) Units are being used, IP Office supports a maximum of:
  - Eight EU24 and or EU24BL per system.
  - Two XM24 units on each Digital Station expansion module to a maximum of 10 XM24 units per system.
  - Two 4450 units on each Digital Station expansion module to a maximum of 10 4450 units per system.
  - T3 DSS units.

See the Telephones Section for specific limits on the number of each type of telephone supported on DS modules.

© 2008 AVAYA All rights reserved.

IP400 Analog Trunk 16 Module

This expansion module provides an additional sixteen Loop Start or Ground Start two-wire analog trunks. (Ground start trunks are not available in all territories) The first two trunks on the module which are automatically switched to power fail sockets on the rear of the unit in the event of power being interrupted must be loop start for correct power fail operation.



© 2008 AVAYA All rights reserved.

© 2008 AVAYA All rights reserved.

Chapter 3. Telephones

© 2008 AVAYA All rights reserved.

3. Telephones

Avaya IP Office Desktop phones are designed to meet typical user needs in the small and medium business market:

- **Walk Up**
  Phones in common areas like a meetings room, lobby or store. The phones are normally used by casual users that walk-up to a phone where they make a quick phone call. Normally, no sophisticated features are needed.
  - **Example user:** Customer, Visitor.
- **Everyday**
  Users that need a phone on their desk as part of their everyday job. While having a phone is important, the phone is just another tool on the desk for users using several alternative communication technologies.
  - **Example user:** Engineer, Accountant.
- **Receptionist/Assistant**
  A typical user manages a lot of calls, often as a significant part of their job. In many cases they also manage calls for others users and need to be efficient in connecting callers to the right people.
  - **Example user:** Executive Assistant, Receptionist.
- **Executive**
  For these users, handling a lot of calls is essential and they are constantly on the phone. Quite often, they handle multiple calls; they must always be connected, even when mobile. They require a set of rich, yet easy to use communication capabilities.
  - **Example user:** Sales Executive, Banker, Lawyer, Project leader.

Phones for the different Usage categories:

- **Walk Up:** 1703, 5602SW/4602SW, 5402/2402
- **Everyday:** 1708, 5610SW/4610SW, 5410/2410
- **Receptionist/Assistant:** 1716 (with BM32 if needed), 5621/4621SW with EU24, 5420/2420 with EU24,
- **Executive:** 5621/4621SW, 5420/2420, 4625SW

© 2008 AVAYA All rights reserved.

1700

Avaya one-X is a portfolio of communications solutions that deliver a powerful and consistent communications experience for the end user – across a variety of devices and interfaces. Avaya one-X solutions provide Intelligent Access to Intelligent Communications driving enhanced productivity and competitive advantage.

Avaya one-X Deskphone Value Edition is a family of cost effective IP Telephones designed especially around the needs of the Small and Medium Business customer.

Designed with the reliability you expect from Avaya, oneX Deskphone Value Edition telephones provide critical features and capabilities not often found in competitively priced models.

The family of 1700 phones consists of three models, designed around different user needs:

- The 1703 phone is designed for the Walk Up user and basic Everyday user. For a Walk Up scenario, the 1703 is used in locations such as common areas in offices, stockrooms, lobbies, or drop-in desks. Users can be building visitors, employees, or even customers who need a phone with a simple, familiar interface.
- The Avaya 1708 IP Telephone is designed for the Everyday user. Everyday users typically rely on several forms of communication including voice and email – and while they require a quality telephone, they rarely handle large amount of calls at the same time. Cubicle workers and sales staff are examples of Everyday users for whom the 1708 delivers a productivity-enhancing telephone.
- The Avaya 1716 IP Telephone is designed for the Receptionist/Assistant user. Receptionists, secretaries, and managers are examples of this type of user – people who answer incoming calls, transfer customers to different departments or extensions, and monitor several line appearances throughout a typical day. For the Receptionist/Assistant user, the 1716 provides the most one-touch line/feature/speed-dial buttons without the need to scroll through on-screen lists.

All 1700 phones share a consistent design line and feature a button-centric layout. With fast access to all needed features like hold, transfer and conference, easily readable back-lit displays and a stylish design, they are an important addition to the Avaya portfolio of phones for the Small and Medium market.

All phones share a number of feature keys with clear, paper labels. On the 1708 and 1716, features are also available through a feature list on screen that makes selecting features and line appearances easy even when hotdesking to another phone.

The phones are accompanied by a complete set of accessories like cost efficient power supplies, easy to use paper labels and spare parts. For powering options refer to the section “Power Options for IP Telephones”. For accessories, refer to the section “Spares”.

For easy and efficient generation of labels, free DESI printing software is available from http://support.avaya.com (on the 1600 Phone Series page). This can be used to print all labels correctly formatted on plain paper or available DESI labels. A PDF document with correct formatting can also be downloaded.

For customer environments requiring Gigabit Ethernet support, a Gigabit Ethernet adapter is available - providing 10/100/1000 Mbps support for both the 1700 series phone and co-located laptop or PC.

Note that with the 1700 series of phones, no Ethernet cable is included. If a cable is needed, it should be ordered separately at the required length.

The 1700 series phones are supported on IP Office from Release 4.2 onwards.

© 2008 AVAYA All rights reserved.

1703 IP Telephone

Works on:	IP Office	Communication Manager
1703	✓	✗

The Avaya 1703 supports 3 line appearances. Each of the buttons includes dual LEDs (red, green) providing explicit status for the user. For a familiar look and feel, the 1703 includes several fixed feature keys for common telephone tasks including conference, transfer, drop, hold, and mute. In addition, the 1703 includes a 2-way speakerphone. The display on the 1703 measures two lines by 16 characters and is backlit for easier viewing in all lighting conditions.

Common Features:

- **Display/Housing:**
  - Backlit display – 2.6" diagonal, 2 rows by 16 characters.
  - Reversible wedge stand for desktop and wall-mount use.
- **Fixed Buttons:** 10.
  - Volume up/down (separate volume levels for the handset, speaker, and ringer), Mute button, Speaker button.
  - Avaya Menu button (options and settings access).
  - Hold button, Conference button, Transfer button, Drop button, Redial button.
- **Programmable/Contextual buttons:** 3.
  - 3 line appearance key buttons – with dual LEDs (red, green) and paper labels (Printing tool available).
- **Key Labels:** Icons used on fixed feature keys.
- **Speakerphone:** Two-way hands-free speaker and microphone.
- **Hearing Aid Compatible:** Yes - ergonomic hearing aid compatible handset also supporting TTD acoustic coupler.
- **Message Waiting Indicator:** Yes - also used as ringing call alert indicator.
- **Personalized Ring Patterns:** Yes - 8.
- **Headset Socket:** Yes.
- **Embedded Applications:**

© 2008 AVAYA All rights reserved.

- Contacts application (100) and Call Log (100, Missed, Incoming, Outgoing).
- Access to company directory and Visual Voice applications on IP Office.

- **Upgradeable Firmware:** Yes.
- **Expansion:** None.
- **Color:** Black.
- **Mounting:** Desk or wall mountable with optional wall mount adapter.

Requirements for IP interface:

- **Power Supply:** Optional IEEE PoE 802.3af adapter (class 2) or optional wall plug local power supply.
- **Codecs/audio:** G.711, G.729a/b, Dynamic Jitter buffer, Echo cancellation, Comfort Noise, Automatic Gain control.
- **QoS Options:** UDP Port Selection, DiffServ and 802.1p/q (VLAN).
- **SNMP Support:** Yes.
- **IP Address Assignment:** Static or dynamic IP address assignment.
- **Ethernet Ports:** Single Ethernet (10/100) line interface.

Language Support:

- Supports the following languages: English, French, Spanish, German, Italian, Dutch, Portuguese and Russian (English and 4 of the other languages at once in the phone)

© 2008 AVAYA All rights reserved.

1708 IP Telephone

Works on:	IP Office	Communication Manager
1708	✓	✗

The Avaya 1708 supports 8 line appearances / feature keys. Each of the buttons includes dual LEDs (red, green) providing explicit status for the user. For a familiar look and feel, the 1708 includes several fixed feature keys for common telephone tasks including conference, transfer, drop, hold, and mute. In addition, the 1708 includes a high quality two-way speakerphone, and supports a broad portfolio of Avaya wired and wireless headsets through its integrated headset jack.

The 1708 features a context sensitive user interface along with three softkeys and a four-way navigation cluster – ideal for scrolling through the local contacts list or call logs. The display on the 1708 measures three lines by 24 characters and is backlit for easier viewing in all lighting conditions.

Common Features:

- **Display/Housing:**
  - Backlit display – 3.5” diagonal, 3 rows by 24 characters.
  - Dual position flip stand.
- **Fixed Buttons:** 15 plus Four-way navigation cluster button.
  - Volume up/down (separate volume levels for the handset, speaker, and ringer), Mute button, Speaker button, Headset button.
  - Avaya Menu button (options and settings access), Telephony application button – to return to main telephone screen.
  - Hold button, Conference button, Transfer button, Drop button.
  - Contacts button, Call log button, Redial button, quick-access voicemail Message button.
- **Programmable/Contextual buttons:** 11.
  - 8 line appearance/feature key buttons – with dual LEDs (red, green) and paper labels (Printing tool available).
  - Three contextual softkey buttons.

© 2008 AVAYA All rights reserved.

- **Key Labels:** Icons used on fixed feature keys.
- **Speakerphone:** High quality two-way hands-free speaker and microphone.
- **Hearing Aid Compatible:** Yes - ergonomic hearing aid compatible handset also supporting TTD acoustic coupler.
- **Message Waiting Indicator:** Yes - also used as ringing call alert indicator.
- **Personalized Ring Patterns:** Yes - 8.
- **Headset Socket:** Yes.
- **Embedded Applications:**
  - Contacts application (100) and Call Log (100, Missed, Incoming, Outgoing).
  - Access to company directory and Visual Voice applications on IP Office.
- **Upgradeable Firmware:** Yes.
- **Expansion:** None.
- **Color:** Black.
- **Mounting:** Desk or wall mountable with optional wall mount adapter.

Requirements for IP interface:

- **Power Supply:** IEEE 802.3af Power over Ethernet (PoE) class 2 device or optional wall plug local power supply.
- **Codecs/audio:** G.711, G.729a/b, Dynamic Jitter buffer, Echo cancellation, Comfort Noise, Automatic Gain control.
- **QoS Options:** UDP Port Selection, DiffServ and 802.1p/q (VLAN).
- **SNMP Support:** Yes.
- **IP Address Assignment:** Static or dynamic IP address assignment.
- **Ethernet Ports:** Ethernet (10/100) line interface with a secondary 10/100 port for collocated laptop or PC with VLAN separation.
  - Well suited for everyday use, where typically smaller sized data files transferred (web, email file attachments) or less demanding data-transfer application/usage (thus less sensitive to data transfer completion times).
  - For highly demanding data traffic processing with large amounts of small packets we recommend the 1716 phone.
  - Support of optional Gigabit Ethernet Adapter.

Language Support:

- Supports the following languages: English, French, Spanish, German, Italian, Dutch, Portuguese and Russian (English and 4 of the other languages at once in the phone)

© 2008 AVAYA All rights reserved.

1716 IP Telephone

Works on:	IP Office	Communication Manager
1716	✓	✗

The Avaya 1716 supports 16 line appearances/feature keys on the phone itself – and up to three 32 button expansion modules, providing access to a total of 112 line/feature keys or speed dial buttons. This will satisfy the needs of even the most demanding secretary or receptionist. Each of the buttons features a dual LED (red, green) providing explicit status for the user. For a familiar look and feel, the 1716 includes several fixed feature keys for common telephone tasks including conference, transfer, drop, hold, and mute. In addition, the 1716 includes a high quality two-way speakerphone, and supports a broad portfolio of Avaya wired and wireless headsets through its integrated headset jack.

The 1716 features a context sensitive user interface along with three softkeys and a four-way navigation cluster – ideal for scrolling through the local contacts list or call logs. The display on the 1716 measures four lines by 24 characters and is adjustable for optimum viewing angle. Additional caller related information is displayed with active appearances for easier call handling. The display is backlit for easier viewing in all lighting conditions.

Common Features:

- **Display/Housing:**
  - Backlit display – 4 rows by 24 characters with adjustable display angle.
  - Dual position flip stand.
- **Fixed Buttons:** 15 plus Four-way navigation cluster button.
  - Volume up/down (separate volume levels for the handset, speaker, and ringer), Mute button, Speaker button, Headset button.
  - Avaya Menu button (options and settings access), Telephony application button – to return to main telephone screen.
  - Hold button, Conference button, Transfer button, Drop button.
  - Contacts button, Call log button, Redial button, Quick-access voicemail Message button.
- **Programmable/Contextual buttons:** 19.

© 2008 AVAYA All rights reserved.

- 16 line appearance/feature key buttons – with dual LEDs (red, green) and paper labels (Printing tool available).
- Three contextual softkey buttons.

- **Key Labels:** Icons used on fixed feature keys.
- **Speakerphone:** High quality two-way hands-free speaker and microphone.
- **Hearing Aid Compatible:** Yes - ergonomic hearing aid compatible handset also supporting TTD acoustic coupler.
- **Message Waiting Indicator:** Yes - also used as ringing call alert indicator.
- **Personalized Ring Patterns:** Yes - 8.
- **Headset Socket:** Yes.
- **Embedded Applications:**
  - Contacts application (100) and Call Log (100, Missed, Incoming, Outgoing).
  - Access to company directory and Visual Voice applications on IP Office.
- **Upgradeable Firmware:** Yes.
- **Expansion:** Module interface – for up to 2 BM32 32-button expansion modules (Note: if more than one BM32 is used an external power supply is needed).
- **Color:** Black.
- **Mounting:** Desk or wall mountable with optional wall mount adapter.

Requirements for IP interface:

- **Power Supply:** IEEE 802.3af Power over Ethernet (PoE) class 2 device or optional wall plug local power supply.
- **Codecs/audio:** G.711, G.729a/b, Dynamic Jitter buffer, Echo cancellation, Comfort Noise, Automatic Gain control.
- **QoS Options:** UDP Port Selection, DiffServ and 802.1p/q (VLAN).
- **SNMP Support:** Yes.
- **IP Address Assignment:** Static or dynamic IP address assignment.
- **Ethernet Ports:** Ethernet (10/100) line interface with a secondary 10/100 port for collocated laptop or PC with VLAN separation.

Language Support:

- Supports the following languages: English, French, Spanish, German, Italian, Dutch, Portuguese and Russian (English and 4 of the other languages at once in the phone).

© 2008 AVAYA All rights reserved.

BM32 Button module.

The BM32 Button Module extends the number of feature buttons on your Avaya 1716 IP Telephone. It provides 32 additional buttons all with dual LEDs (red, green) for easy visibility.

Up to 2 button modules can be connected to a single 1716 phone. The total number of button modules supported on one IP Office system is 16. Note that appearance buttons are not supported on a BM32.

Typical use cases are:

- to provide additional lines for calls bridged from one or more other extensions
- to provide abbreviated dialing buttons
- to provide buttons to access additional switch features like call forwarding

One button module can be supported without an additional power supply on a 1716 phone. If 2 button modules are needed a separate external power supply must be connected to the 1716 phone.

© 2008 AVAYA All rights reserved.

4601 Telephone

Works on:	IP Office	Communication Manager
5601*	✓	✗
4601	✓	✓

*Note the 5601 is no longer sold by Avaya but is supported on IP Office Release 4.2.

Common Features:

- **Display:** None.
- **Fixed Feature Buttons:** 8 - Conference, Transfer, Drop, Redial, Messages, Hold, Volume Up, Volume Down.
- **Programmable Feature Buttons:** 2 with single color indicator lamps.
- **Key Labels:** Icons used on fixed feature keys. None on programmable feature keys.
- **Speakerphone:** No.
- **Hearing Aid Compatible:** Yes.
- **Message Waiting Indicator:** Yes.
- **Personalized Ring Patterns:** No.
- **Headset Socket:** No, this phone does not support headset operation.
- **Embedded Applications:** None.
- **Upgradeable Firmware:** Yes.
- **Expansion:** None.
- **Color:** Multi-gray.
- **Power Supply:** IEEE 802.11af Power over Ethernet (PoE) or individual Avaya power supply unit (1151).
- **Connect to:** LAN using H.323 VoIP.
- **Mounting:** Desk or wall mountable.
- **Adjustable Desk Stand:** No.
- **Codecs:** G.711, G.729a/b.
- **QoS Options:** UDP Port Selection, DiffServ and 802.1p/B (VLAN)
- **SNMP Support:** Yes.
- **IP Address Assignment:** Dynamic IP address assignment only
- **Ethernet Ports:** Single 10/100 BaseT Ethernet port.

© 2008 AVAYA All rights reserved.

5402, 5602 SW, 2402, 4602 SW Telephone

Telephone	Works on IP Office	Works on IP Office and Communication Manager
5402	✓	✗
5602 SW	✓	✗
2402	✓*	✓
4602 SW	✓	✓

*Early 2402 telephones can make and receive call but the display will not function.

Common Features:

- **Display:** 2 lines x 24 characters.
- **Fixed Feature Buttons:** 10 - Conference, Transfer, Drop, Redial, Speaker, Messages, Hold, Mute, Volume Up, Volume Down.
- **Programmable Feature Buttons:**
  - **DS Phones:** 2 plus an additional 12 programmable feature keys can be accessed via the **FEATURE** key.
  - **IP Phones:** 2.
- **Key Labels:** Icons used on fixed feature keys. Display labels and icons used on 2 programmable feature keys.
- **Speakerphone:** Listen-only hands free speaker (no microphone).
- **Hearing Aid Compatible:** Yes.
- **Message Waiting Indicator:** Yes. On the 2402 and 5402 this is also used as a ringing call alert indicator.
- **Personalized Ring Patterns:** Yes- 8 (via Conference button when idle).
- **Headset Socket:** No, this telephone does not support headset operation.
- **Embedded Applications:** None.
- **Upgradeable Firmware:** DS Phones - No. IP Phones - Yes.
- **Expansion:** None.
- **Color:** Multi-gray.
- **Mounting:** Desk or wall mountable.
- **Adjustable Desk Stand:** No.

© 2008 AVAYA All rights reserved.

Requirements for 5402 and 2402:

- Connect to: Digital Station (DS) port.
- Power Supply: From phone system.

Requirements for 5602 SW and 4602 SW:

- **Power Supply:** IEEE 802.3af Power over Ethernet (PoE) or individual power supply unit (Avaya 1151 series).
- **Codecs:** G.711, G.729a/b.
- **QoS Options:** UDP Port Selection, DiffServ and 802.1p/q (VLAN)
- **SNMP Support:** Yes.
- **IP Address Assignment:** Static or dynamic IP address assignment.
- **Ethernet Ports:** Two port full-duplex 10/100 BaseT Ethernet switch for PC pass-through connection.
  - Auto-negotiation provided separately for each port.
  - 802.3 Flow Control.
  - Phone has priority over PC port at all times.

© 2008 AVAYA All rights reserved.

5410, 5610 SW, 2410, 4610 SW Telephones

Telephone	Works on IP Office	Works on IP Office and Communication Manager
5410	✓	✗
5610 SW	✓	✗
2410	✓	✓
4610 SW	✓	✓

Common Features:

- **Display:** 5 lines x 29 characters (168 x 80 pixel 4-grayscale).
- **Fixed Feature Buttons:** 10 - Conference, Headset, Transfer, Drop, Redial, Speaker, Hold, Mute, Volume Up, Volume Down.
- **Programmable Feature Buttons:**
  - **DS Phones:** 12 - in 2 switchable display pages of 6 matching the 6 physical display buttons.
  - **IP Phones:** 24 - in 4 switchable display pages of 6 matching the 6 physical display buttons.
- **Key Labels:** Icons used on fixed feature keys.
- **Speakerphone:** Two-way hands-free speaker and microphone.
- **Hearing Aid Compatible:** Yes.
- **Message Waiting Indicator:** Yes - also used as ringing call alert indicator.
- **Personalized Ring Patterns:** Yes - 8.
- **Headset Socket:** Yes.
- **Embedded Applications:** Speed Dial List (48) and Call Log (Missed, Incoming, Outgoing). Also WAP WML browser supported on IP phone models.
- **Upgradeable Firmware:** Yes.
- **Expansion:** None.
- **Color:** Multi-gray.
- **Mounting:** Desk or wall mountable.
- **Adjustable Desk Stand:** Yes - Supplied with phone.

© 2008 AVAYA All rights reserved.

Special Features for the 5410 and 2410:

- **Messages Button:** Dedicated button to collect voicemail.

Requirements for 5410 and 2410:

- **Connect to:** Digital Station (DS) port.
- **Power Supply:** From phone system.

Requirements for 5610 and 4610:

- **Power Supply:** IEEE 802.3af Power over Ethernet (PoE) or individual power supply unit (Avaya 1151 series).
- **Codecs:** G.711, G.729a/b.
- **QoS Options:** UDP Port Selection, DiffServ and 802.1p/q (VLAN)
- **SNMP Support:** Yes.
- **IP Address Assignment:** Static or dynamic IP address assignment.
- **Ethernet Ports:** Two port full-duplex 10/100 BaseT Ethernet switch for PC pass-through connection.
  - Auto-negotiation provided separately for each port.
  - 802.3 Flow Control.
  - Phone has priority over PC port at all times.

© 2008 AVAYA All rights reserved.

5420, 5621, 2420, 4621, 4625 Telephones

Telephone	Works on IP Office	Works on IP Office and Communication Manager
5420	✓	✗
5621 SW	✓	✗
2420	✓	✓
4621 SW	✓	✓
4625 SW	✓	✓

Common Features:

- **Display:** 7 lines x 29 characters.
- **Fixed Feature Buttons:** 10 - Conference, Headset, Transfer, Drop, Redial, Speaker, Hold, Mute, Volume Up, Volume Down.
- **Programmable Feature Buttons:**
  - **DS Phones:** 24 - arranged in 3 switchable display pages of 8 matching the 8 physical display buttons.
  - **IP Phones:** 24 - arranged in 2 switchable display pages of 12 matching the 12 physical display buttons.
- **Key Labels:** Icons used on fixed feature keys.
- **Speakerphone:** Two-way hands free speaker and microphone.
- **Hearing Aid Compatible:** Yes.
- **Message Waiting Indicator:** Yes - also used as ringing call alert indicator.
- **Personalized Ring Patterns:** Yes - 8.
- **Headset Socket:** Yes.
- **Embedded Applications:** Speed Dial List (104) and Call Log (Missed, Incoming, Outgoing). Also WAP WML browser supported on IP phone models.
- **Upgradeable Firmware:** Yes.
- **Expansion:** Supports the EU24 DSS expansion module (with additional Avaya 1151 power supply).
- **Color:** Multi-gray.
- **Mounting:** Desk or wall mountable.
- **Adjustable Desk Stand:** Yes - Supplied with phone.

© 2008 AVAYA All rights reserved.

Special Features for the 5420 and 2420:

- **Messages Button:** Dedicated button to collect voicemail.

Requirements for 5420 and 2420:

- **Connect to:** Digital Station (DS) port.
- **Power Supply:** From phone system.

Requirements for 5621 SW, 4621 SW, 4625 SW:

- **Power Supply:** IEEE 802.3af Power over Ethernet (PoE) or individual power supply unit (Avaya 1151 series).
- **Codecs:** G.711, G.729a/b.
- **QoS Options:** UDP Port Selection, DiffServ and 802.1p/q (VLAN)
- **SNMP Support:** Yes.
- **IP Address Assignment:** Static or dynamic IP address assignment.
- **Ethernet Ports:** Two port full-duplex 10/100 BaseT Ethernet switch for PC pass-through connection.
  - Auto-negotiation provided separately for each port.
  - 802.3 Flow Control.
  - Phone has priority over PC port at all times.

Special Features for 5621 SW and 4621 SW:

- **Display Backlight:** The display has a backlight for improved contrast. Standby mode turns off backlight after time-out.

Special Features for 4625 SW:

- **Color Backlight Display:** The display is full color and has a backlight for improved contrast.

Note: While still supported, the 5620SW and 4620SW phones are no longer available for purchase.

© 2008 AVAYA All rights reserved.

EU24 and EU24 BL Expansion Modules

The EU24/EU24BL are phone expansion modules that work in association with a 5420, 5620/1, 2420, 4620/1, 4625 phones. They provide an additional 24 programmable buttons with associated display label and status icons. Only one EU24 can be used per phone. The EU24BL has a backlight and is for use with the 4621 and 5621 only.

The EU24/EU24BL supports an additional 24 Call Appearance/Feature buttons, by displaying a column of 12 buttons at a time, with a dotted line separating the two columns.

Common Features

- 24 Programmable call appearance/feature keys.
- Automatically labeled from the system (no paper labels).
- Connects directly to the associated phone.
- Requires an Avaya 1151 series power supply, even for IP phones already using Power over Ethernet (PoE).
- IP Office supports a maximum of eight EU24/EU24 BL's on each IP Office system.

Telephone	EU24	EU24BL
2402/5402	✗	✗
2410/5410	✗	✗
2420/5420	✓	✗
4601/5601	✗	✗
4602/5602	✗	✗
4610/5610	✗	✗
4620/5620	✓	✗
4621/5621	✓	✓
4625	✓	✓

© 2008 AVAYA All rights reserved.

T3 Telephone Range

The T3 range of digital Upn and IP telephones provide European style with context sensitive displays and are available in select European countries only.

© 2008 AVAYA All rights reserved.

T3 Compact

Common Features:

- **Display:** 1 line with 24 characters alphanumerical plus one line icons
- **Fixed Feature Buttons:** 3 keys with printed text labels and 2 for Audio Volume control
- **Programmable Feature Buttons:** 3 keys with indicators and printed text labels, 2 keys with printed text labels
- **Speakerphone:** Full duplex hands free speaker and microphone.
- **Hearing Aid Compatible:** Through optional handset
- **Message Waiting and call log Indicator:** Yes
- **Personalized Ring Patterns:** Yes, 8 ring patterns
- **Embedded Applications:** Navigation Cursor Control, Call signaling via LED and/or ringer; Alphanumeric entry via dialing keypad.
- **Color:** Graphite gray or polar white.
- **Mounting:** Desk or optional wall mountable.
- **Adjustable Desk Stand:** No

Features for T3 Upn only:

- **Upgradeable Firmware:** No.
- **Optional Add-Ons:** up to 3 DSS Modules, T3 Headset link for wired headsets only
- **Headset Socket:** No
- **Connect to:** Digital Station (DS) port.
- **Power Supply:** From phone system.

Features for T3 IP only:

- **Upgradeable Firmware:** Yes
- **Headset Socket:** Yes
- **Optional Add-Ons:** No
- **Power Supply:** IEEE 802.3af Power over Ethernet (PoE) or individual power supply unit.
- **Codecs:** G.711, G.729a/b.
- **QoS Options:** UDP Port Selection, DiffServ and 802.1p/q (VLAN)
- **SNMP Support:** No.
- **IP Address Assignment:** Static or dynamic IP address assignment.
- **Ethernet Ports:** Two port full-duplex 10/100 BaseT Ethernet switch.
  - Auto-negotiation provided separately for each port.
  - 802.3 Flow Control.

© 2008 AVAYA All rights reserved.

T3 Classic

Common Features:

- **Display:** graphical, 4 lines x 26 characters
- **Fixed Feature Buttons:** 5 preprogrammed keys with printed text labels and 2 for Audio Volume control
- **Programmable Feature Buttons:** 6 preprogrammed keys with indicators and printed text labels, 4 programmable keys with printed text labels
- **Speakerphone:** Two-way hands free speaker and microphone.
- **Hearing Aid Compatible:** Through optional handset
- **Message Waiting and call log Indicator:** Yes
- **Personalized Ring Patterns:** Yes, 8 ring patterns.
- **Headset Socket:** no
- **Embedded Applications:** Navigation Cursor Control, Call signaling via LED and/or ringer; Alpha entry via dialing keypad.
- **Optional Add-Ons:** T3 Headset link for wired headsets only
- **Color:** Graphite gray or polar white.
- **Mounting:** Desk
- **Adjustable Desk Stand:** Display adjustable

Features for T3 Upn:

- **Upgradeable Firmware:** No.
- **Optional Add-Ons:** up to 3 DSS Modules
- **Connect to:** Digital Station (DS) port.
- **Power Supply:** From phone system.

Features for T3 IP:

- **Upgradeable Firmware:** Yes.
- **Optional Add-Ons:** up to 3 DSS Modules with AEI/Headsetlink,
- **Power Supply:** IEEE 802.3af Power over Ethernet (PoE) or individual power supply unit.
- **Codecs:** G.711, G.729a/b.
- **QoS Options:** UDP Port Selection, DiffServ and 802.1p/q (VLAN)
- **SNMP Support:** no.
- **IP Address Assignment:** Static or dynamic IP address assignment.
- **Ethernet Ports:** Two port full-duplex 10/100 BaseT Ethernet switch.
  - Auto-negotiation provided separately for each port.
  - 802.3 Flow Control.

© 2008 AVAYA All rights reserved.

T3 Comfort

Common Features:

- **Display:** graphical 17 lines x 40 characters, Integrated keyboard
- **Fixed Feature Buttons:** 5 preprogrammed keys with printed text labels and 2 for Audio Volume control
- **Programmable Feature Buttons:** 6 preprogrammed keys with indicators and printed text labels, 6 preprogrammed keys with printed text labels, 10 user programmable keys with associated display labels.
- **Speakerphone:** Two-way hands free speaker and microphone.
- **Hearing Aid Compatible:** Through optional handset
- **Message Waiting and call log Indicator:** Yes
- **Personalized Ring Patterns:** Yes, 8 ring patterns.
- **Headset Socket:** No
- **Embedded Applications:** Navigation Cursor Control, Call signaling via LED and/or ringer
- **Optional Add-Ons:** T3 Headset link for wired headsets only
- **Color:** Graphite gray or polar white.
- **Mounting:** Desk
- **Adjustable Desk Stand:** Display adjustable

Features for T3 Upn:

- **Upgradeable Firmware:** No.
- **Optional Add-Ons:** up to 3 DSS Modules
- **Connect to:** Digital Station (DS) port.
- **Power Supply:** From phone system.

Features for T3 IP:

- **Upgradeable Firmware:** Yes.
- **Optional Add-Ons:** up to 3 DSS Modules with AEI/Headsetlink,
- **Power Supply:** IEEE 802.3af Power over Ethernet (PoE) or individual power supply unit.
- **Codecs:** G.711, G.729a/b.
- **QoS Options:** UDP Port Selection, DiffServ and 802.1p/q (VLAN)
- **SNMP Support:** No.
- **IP Address Assignment:** Static or dynamic IP address assignment.
- **Ethernet Ports:** Two port full-duplex 10/100 BaseT Ethernet switch.
  - Auto-negotiation provided separately for each port.
  - 802.3 Flow Control.

© 2008 AVAYA All rights reserved.

T3 DSS Expansion Modules

The T3 DSS Module is a phone expansion module that is compatible with all T3 Upn and T3 IP Telephones except the T3 IP Compact. Each module provides an additional 36 programmable buttons with associated printed text labels and indicators, and can be programmed for lines, groups or speed dial numbers. 3 DSS Modules can be added to each T3 phone. Power is provided by T3 Upn telephones, but an external power adapter is needed for each DSS module when used on T3 IP telephones.

IP Office 406, IP412 and IP500 support a maximum of 30 T3 DSS modules per control unit.

© 2008 AVAYA All rights reserved.

Wireless and Remote Phone Solutions

Avaya IP Office Wireless Solutions include analog, digital and IP-based WiFi wireless phones. These are solutions employees can use every day to work more effectively and be more responsive to customers — all while increasing revenues and keeping communication costs firmly under control. Also, Avaya IP Office Mobility Solutions integrate seamlessly with IP Office, enhancing each customer's investment. IP Office's in-building Mobility Solutions improve communication with staff that, because of the function they perform, are mobile within the workplace. Using wireless technology, such individuals may be instantly contactable, with many obvious benefits;

- The wireless telephone is carried in the pocket, so users are not tied to the desk in order to remain in contact.
- Users may be contacted instantly to ensure fast, accurate decision making and immediate response to problems through planned radio coverage with no blind spots

Avaya Wireless and Remote Solutions

IP Office supports the following wireless solutions:

- DECT in the EMEA and NA regions and in selected APAC countries.
- Digital Wireless North American market.
- Avaya VoIP WiFi Solution offered worldwide in selected countries including North America, EU countries and selected others.

IP Office supports the following VPN remote phone solutions:

- VPN phone client on 4610SW, 4621SW, 5610SW and 5621SW IP phones offered worldwide.

© 2008 AVAYA All rights reserved.

Wireless - Avaya IP DECT

The IP DECT solution delivers the productivity-boosting benefits of IP and wireless communications across multiple offices in a convenient, lightweight handset. It provides businesses with a highly functional wireless solution with the ability to scale to support large numbers of users. This system also supports users in different offices connected via a WAN. The Avaya IP DECT solution radio fixed part (RFP) or base station connects to the IP Office using an IP protocol based on H.323.

The Avaya IP DECT solution supports up to 120 handsets and 32 base stations. Each base station can be powered over the LAN using the Power over Ethernet (PoE) standard. Each indoor base station can also optionally be connected to main power via an external power adaptor. Each outdoor base station can only be powered using PoE - no individual power supplies are available to power the outdoor IP DECT base station.

In EMEA and APAC this system supports the 3701 and 3711 handsets.

In North America, only the 3711 handset is supported.

Avaya recommends that for new deployments, for full feature functionality the 3711 handset be used with the IP DECT solution.

Note: The regulatory requirements for the radio part (base station and Handset) are slightly different in the US and Canada compared to EMEA and APAC. Therefore, while providing the same functionality, the hardware is different in these two regions.

Each Base station has the following features:

- 8 simultaneous Voice and up to 12 Signaling Channels.
- Codec G.711, G.723, G.729 for base station IP trunk connection.
- Handover

  While in motion, the handset performs continuous measurements to determine which IP DECT base station has the strongest signal. The one that can be best received is defined as the active Base station. To prevent the handset from rapidly switching back and forth between two base stations that are equally well received, threshold values are used. Handover between base stations occurs seamlessly whether a call is active or not.
- DECT Networking

  An IP DECT telephone can travel from one office to another which is connected over a wide area network (WAN) link and make and take calls. In this scenario the main IP DECT controller remains at one "headquarters" location.

Given the degree of integration available to wireless users with DECT, there are a variety of means by which calls can be routed to wireless handsets:

- **DDI/DID**
  Since each wireless handset is an extension on the IP Office system calls may be routed directly using a DDI/DID number.
- **Transfer**
  Calls may be transferred to DECT extensions by operators or other extension users and DECT extension users may transfer callers to any other extension user.
- **Hunt Group compatibility**
  Wireless handsets may be programmed as members of groups and answer calls in the same manner as any other extension within that group.
- **Group working**
  Wireless handsets may be programmed as members of groups and attract calls in the same manner as any other extension within that group. DECT handsets must NOT be configured into collective groups.
- **Divert destination**
  Users may initiate any or all diverts from an Avaya desk phone to a wireless handset.
- **Twinning**
  Twinning allows calls to a user main extension number to alert at both that extension and a secondary extension. Though not restricted to DECT, this feature is aimed primarily at users who have both a desk phone and a wireless extension. Calls from the secondary twinned extension are presented as if from the users main extension. Presentation of call waiting and busy is based on whether either of the twinned extensions is in use. In North America this functionality became available in Release 4.0.7.

© 2008 AVAYA All rights reserved.

Avaya IP DECT System licensing

A license is necessary for this functionality. This license is called the Avaya IP Office IP DECT Mobility Manager license. This license is entered through the main base station (ADMM) and is NOT entered through the IP Office System manager. A feature key server is NOT necessary to enable the IP DECT functionality.
No separate PC or software is required with this system.

In all regions, a "plug and play" licensing mechanism is available: It consists of a pre-licensed and ready to go two-base-station bundle ("IP DECT IPO STARTER KIT ") and two pre-licensed base stations ("IP DECT RFP32/34 UPG KIT") that can be added to the system independent of the number of licenses in ADMM. This allows easier deployment and upgrades of systems without the need to buy a separate upgrade-license. For IP Office we recommend to use the "Starter Kits" and the "Upgrade Kits" for new installations for added flexibility and to minimize the installation effort.

The bundles that have previously been available in EMEA will continue to be available and are compatible with the pre-licensed base stations described above, if the latest software is installed on the IP-DECT system, e.g. upgrade an existing 5-base-station system with an "Upgrade-Kit" when adding an extra base-station instead of upgrading the system-license to a 6+ license.

Additional upgrade licenses will continue to be available for systems that need to expand their current coverage or capacity.

IP DECT Capacities

Feature	IP DECT
Maximum handsets	120
Maximum base-stations	32
Total base-stations/repeaters	32
Maximum simultaneous calls	100*

*May be limited by the available VCM voice compression channels for calls to non-IP destinations.

© 2008 AVAYA All rights reserved.

Wireless - 900MHz Digital Wireless

The Avaya Digital Wireless uses the 902 to 928 MHz ISM (Industrial, Scientific, and Medical) band. Unlike some other in-building wireless systems, there are no airtime charges, and no license is required. This handset uses digital radio technology and spread-spectrum frequency hopping to provide extremely secure wireless communications.

The Avaya 3810 wireless telephone is a digital telephone designed to work with IP Office (minimum release 2.0). It offers the mobility inherent in a wireless telephone plus access to a number of features and functionality of the connected communications system. The Avaya 3810 wireless telephone uses 900 MHz digital technology allowing a maximum range of 160 feet from the base station.

A maximum of 5 Avaya 3810 wireless handsets can be used in the same zone of radio coverage, Site Planning rules do apply, please refer to installation guide available from the following web site: http://www.avaya.com/support and then select

- Product Documentation
- Telephone Devices and User Agents

Full documentation is also contained within the package.

© 2008 AVAYA All rights reserved.

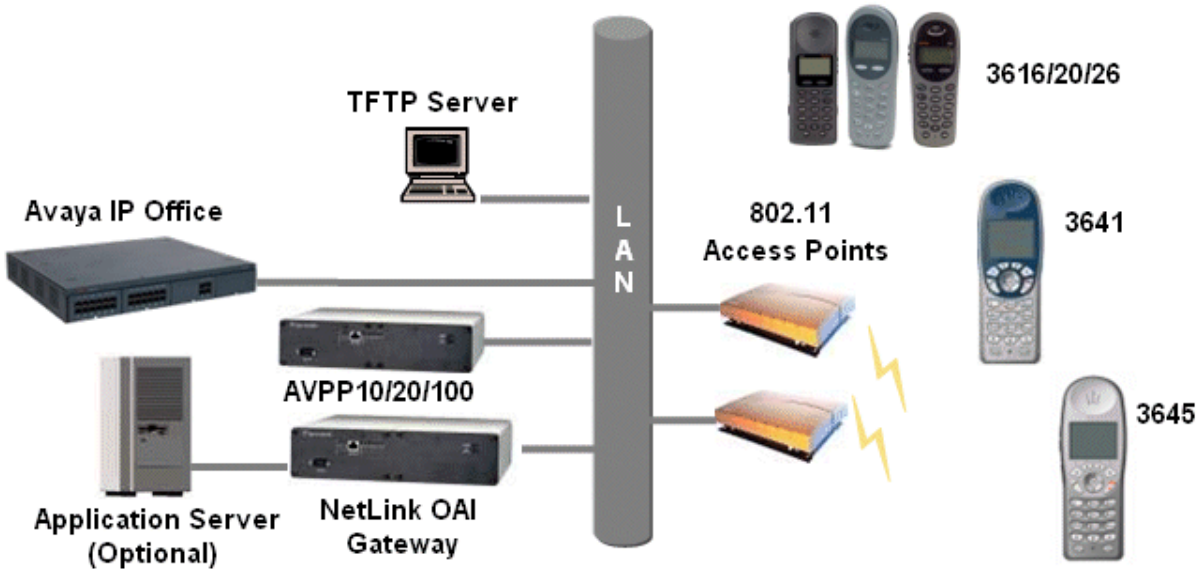
Wireless - WiFi (802.11)

The Avaya IP Wireless solution offers an advanced Voice over IP (VoIP) client for wireless networks. This solution allows SME's to take advantage of the cost savings and simplified management of a converged voice and data infrastructure.

Avaya 3616, 3620 and 3626 phones are optimized for Avaya IP telephony and emulate the wired 4606 IP Telephone. They work in conjunction with the Avaya Voice Priority Processors (10, 20 and 100) to ensure voice quality over Wireless LAN's.

The newly available 3641 and 3645 phones provide an improved user-interface, a new lightweight design and a radio that supports several WIFI standards (802.11a/b/g). With these handsets customers have an increased choice to fit their needs and infrastructure.

Based on global standards for wireless LAN's, the Avaya IP Wireless Telephone Solution simplifies network infrastructure by enabling voice traffic to be carried along with data traffic over the same wireless network. 3616, 3620 and 3626 telephones are available for direct sequence 802.11b Wi-Fi networks; the 3641 and 3645 will also work in 802.11a and 802.11g networks. These phones are also field upgradeable through external TFTP clients (not included), so telephones can be updated with new protocols, features, and capabilities as they become available.



Users can have a choice of an executive or rugged workplace telephone and all the productivity benefits of their desk telephone in this next generation of wireless telephone solutions.

© 2008 AVAYA All rights reserved.

Wireless IP Terminals

Users can have a choice of 5 WiFi phones to meet their in building mobility needs:

- 3 Phones supporting the 802.11b standard and up to 6 lines or selected features:
  - Avaya 3616 supports a broad range of enterprise applications and is ideally suited for general office, financial or hospitality industries. This compact handset offers a high-resolution graphic display and menu driven functions.
  - Avaya 3620 is specifically designed to meet the needs of the healthcare vertical. The 3620 comes standard with a backlit display.
  - Avaya 3626 is an extremely durable handset for workplace applications in industrial environments. This phone is easy to use and requires minimal training. Push-to-talk functionality is also available for broadcast communication between employees, eliminating the need for two-way radios or walkie talkies. The large ear piece seals out background noise and provides comfort for frequent or lengthy calls.
  - Access to 6 lines or selected features through the "line" key (no labels).
- 2 Phones supporting the 802.11 a/b/g standard and up to 12 lines or selected features. Both of these handsets are resistant to dust and spraying water and therefore also suitable for harsh environments. They also offer office-quality speaker-phone functionality.
  - Avaya 3641 supports a broad range of enterprise applications and is ideally suited for general office, financial or hospitality industries. This compact handset offers a high-resolution backlight graphic display a new, improved user-interface and design and a lightweight form factor.
  - Avaya 3645 is a slightly larger version that in addition supports “push-to-talk” functionality for broadcast communication between employees. Due to its rubberized sized grips and the larger ear cup it is especially well suited in noisy and industrial environments.
  - Access to 2*6 lines or selected features through the "line" key and the "FCN" key (no labels).

© 2008 AVAYA All rights reserved.

Avaya Voice Priority Processors

The Avaya Voice Priority Processor (AVPP) is an Ethernet LAN appliance that works with access points to provide Quality of Service (QoS) on the wireless LAN. All packets to and from the wireless phones pass through the AVPP and are encapsulated for prioritization as they are routed to and from IP Office. AVPP is fully compliant with the IEEE 802.11a/b/g standards.

AVPP is required for QoS because the current IEEE 802.11a/b/g wireless LAN standards provide only limited mechanism for differentiating audio packets from data packets. It also delivers quality of service by limiting the number of phones that are connected to one access point in order to avoid quality problems. In addition AVPP ensures that the phone can run in energy-efficient mode when not in use. The following AVPPs are available to meet customer needs:

- AVPP100: Serves 80 calls simultaneously.
- AVPP020: Serves 20 powered-on handsets.
- AVPP010: Serves 10 powered-on handsets.

Wireless Access Points

When using the Avaya Wireless IP solution, customers can utilize wireless access points from various vendors. The list of compatible wireless access points is large and constantly growing. Please visit http://www.spectralink.com/consumer/support/index.jsp and select "WLAN Compatibility List" for the latest information.

Benefits

- Supports the 802.11b or 802.11 a/b/g standard for Wi-Fi networks converging voice and data over a single network.
- Seamless integration with IP Office.
- Excellent voice quality on converged wireless networks.
- Lightweight, durable handsets specifically designed for workplace use.
- Improved display, battery life, processor power all with lower costs.
- Increased range of AVPP's to address the needs of diverse construct sizes.
- Multitude of accessories are available:
  - Dual Charger (full charge accomplished in approximately one and a half hours).
  - Quick Charger (full charge accomplished in approximately one and a half hours).
  - Single, Dual, and Quad Chargers for the 3641 and 3645 phones.
  - Belt Clip.
  - Nylon Pouch.
  - Carrying case with Lanyard.
  - Hands Free Pouch.
  - Noise canceling headset.
  - Over the ear headset.

Avaya IP Wireless Telephony Solution (AWTS) Open Application Interface (OAI) Gateway

The AWTS Open Application Interface (OAI) Gateway enables third- party software applications to communicate with the Avaya IP Wireless Telephones. This serves as a two-way messaging device. Many companies provide applications that interface to your in-house paging systems, email, and client-server messaging. Other vendors with complementary systems such as nurse call, telemetry, alarm, and control system manufacturers are currently developing applications to interface with the Avaya IP Wireless Telephone solution.

© 2008 AVAYA All rights reserved.

3616 Wireless Telephone

The Avaya 3616 IP Wireless Telephone is a WiFi (802.11b) telephone that runs using H.323.

The 3616 supports the following features:

- Lightweight innovative design .
- Simple to use.
- 802.11b standard-compatible.
- Radio Frequency 2.4000 – 2.835 GHz (SMI).
- Transmission type Direct Sequence Spread Spectrum (DSSS).
- FCC certification Part 15.247.
- Management of telephones via DHCP and TFTP.
- Voice encoding G711.
- Transmit Power 100mw peak, <10mW average.
- Wired Equivalent Privacy (WEP), 40bit and 128 bit.
- 2x16 character alphanumeric, plus status indicators.
- 4 hours talk time and 80 hours standby.

© 2008 AVAYA All rights reserved.

3620 Healthcare Wireless Telephone

The Avaya 3620 IP Wireless Telephone is a WiFi (802.11b) telephone that runs using H.323.

The 3620 supports all of the features of 3616 with the following differences:

- Designed for health care environments
- Waterproof durable design.
- Display Backlight:
- Manufacturer’s Liquid damage warranty

© 2008 AVAYA All rights reserved.

3626 Ruggedized Wireless Telephone

The Avaya 3626 Wireless Telephone is a WiFi standard (802.11b) telephone that runs using H.323.

The 3626 supports all of the features of 3616 with the following differences:

- Designed for industrial environments.
- Ruggedized durable design.
- Push-to-talk (walkie-talkie) feature for broadcast communications between employees.

Note: 3626 supports both R1.0 and R2.0 firmware on the set itself. However, as of R3.1 of IP Office, only 3626 phone R1.0 firmware is supported.

© 2008 AVAYA All rights reserved.

3641 Ruggedized Wireless Telephone

The Avaya 3641 Wireless Telephone is a WiFi standard (802.11a/b/g) telephone that runs using H.323.

The 3641 supports the following features:

- Slim lightweight design with large display.
- Backlight display with Icons.
- Simple to use with improved user interface.
- Navigation and soft keys for simple access to frequently used operations.
- Office-quality speakerphone for hands-free operation.
- 802.11a/b/g standard-compatible.
- Radio Frequency 2.4000 GHz (b/g) or 5.8 GHz (a).
- FCC certification Part 15.247.
- Management of telephones via DHCP and TFTP.
- Voice encoding G711, G.729a.
- Wired Equivalent Privacy (WEP), 40bit and 128 bit and 802.11i (PSK) for secure communication.
- Lithium Ion Battery pack with up to 8 hours talk time and 160 hours standby.
- IP-53 Design (Liquid/dust protection).
- MIL 810F Design (Shock protection).
- Clips, cases, lanyard.

© 2008 AVAYA All rights reserved.

3645 Ruggedized Wireless Telephone

The Avaya 3645 Wireless Telephone is a WiFi standard (802.11a/b/g) telephone that runs using H.323.

The 3645 supports all of the features of 3641 with the following additions:

- Push-to-talk (PTT) functionality for workgroup communication
- Enlarged earpiece for operation in noisy environments
- Rubberized grips for improved ergonomics and durability

© 2008 AVAYA All rights reserved.

3701 IP DECT

This handset is supported on the Avaya IP DECT system only. This handset is not available in North America.

- Listen-only hands free speaker.
- SOS Emergency key for speed dialing an emergency number.
- Information key that can be used for:
  - Phone number lists and voice mail indication.
  - Information and speaker key flash when active.
- 50 phone book entries in every handset
- 10 possible ring tones with temporary mute.
- 4-level signal strength display.
- Speaker and handset volume, 3-levels and mute capability.
- Manual and automatic key lock (1 minute timer).
- Temporary ring tone muting.
- Silent charging.
- 12 menu languages: Czech, Danish, Dutch, English, Finnish, French, German, Italian, Norwegian, Portuguese, Spanish and Swedish. However, in the Czech and Norwegian language mode some menu items may appear in the English language.
- Illuminated 3-line graphic display (96 x 33 pixels), variable 3-level contrast.
- Stand-by time: up to 200 hours.
- Talk time: up to 20 hours.
- Charge time: max. 6 hours for empty batteries.
- Weight: 138 grammes including 3 AAA (NiMH) batteries.
- Dimensions (Height x Width X Depth): 146 x 55 x 28 mm.

Optional telephone accessories include:

- Desktop charger.
- An adapter cord for use with headsets.
- Heavy-duty belt clip.

© 2008 AVAYA All rights reserved.

3711 IP DECT

This telephone is supported on the Avaya IP DECT system only.

The 3711 phone supports the same features as the 3701 IP DECT handset but with the following differences:

- Full hands-free speakerphone operation.
- Headset connection (2.5 mm jack).
- Vibrating alarm.
- Personal phone book with 100 entries
- Access to system phone book.
- Voice Mail indication.
- Choice from 30 ring tones.
- Speaker and handset volume, 7-levels and mute capability.
- Automatic call pick-up using a headset.
- 10 menu languages: Danish, Dutch, English, Finnish, French, German, Italian, Portuguese, Spanish and Swedish.
- Illuminated 5-line graphic display, (96 x 60 pixels), variable 7-level contrast.

Optional handset accessories include:

- Desktop charger.
- An adapter cord for use with headsets.
- Heavy-duty belt clip.

© 2008 AVAYA All rights reserved.

Digital Wireless 3810 Telephone

Features

- 2-line, 32 character Handset Liquid Crystal Display (LCD).
- 10 hours of talk time, and 4 days of standby time.
- 4 displayed operation modes indicating Talk, Ringer On/Off, Battery Low, and Message Waiting.
- Single button access to fixed features – Hold, Transfer, Conference, and Redial.
- 4 programmable buttons to access features on the PBX.
- 20 Number Memory for quick and easy speed dialing
- 10 channels, supporting up to 10 simultaneous conversations in overlapping radio coverage areas.
- Headset jack.
- Ringer and Handset volume control.
- User selectable ring type.
- Vibrate alert.
- Redial Button
- Base Unit and Charger Unit.

The Avaya 3810 Wireless Telephone is a digital telephone designed to work with IP Office from release 2.0 and above by connecting to a Digital Station (DS) port. It offers the mobility inherent in a wireless telephone plus access to a number of features and functionality of the connected communications system.

A maximum of 5 Avaya 3810 wireless handsets can be connected to the same IP Office in any overlapping radio coverage area.

This phone is only available in the USA and Canada.

© 2008 AVAYA All rights reserved.

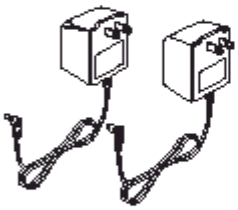
The Avaya 3810 is delivered as a single unit containing:

- Base Unit.
- Handset.
- Telephone Cord.
- Base Unit Power Supply Adapter.
- Charging Stand Power Supply Adapter.
- Rechargeable Battery.
- Belt Clip.
- Charging Stand.
- User and Installation Guide.
- Wall Plate Adapter.

Base Unit

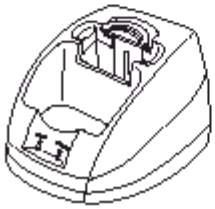
Handset

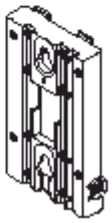
Telephone Cord


2 AC Adapters

Rechargeable Battery

Beltclip


Charger Unit


Wall Mount Stand

© 2008 AVAYA All rights reserved.

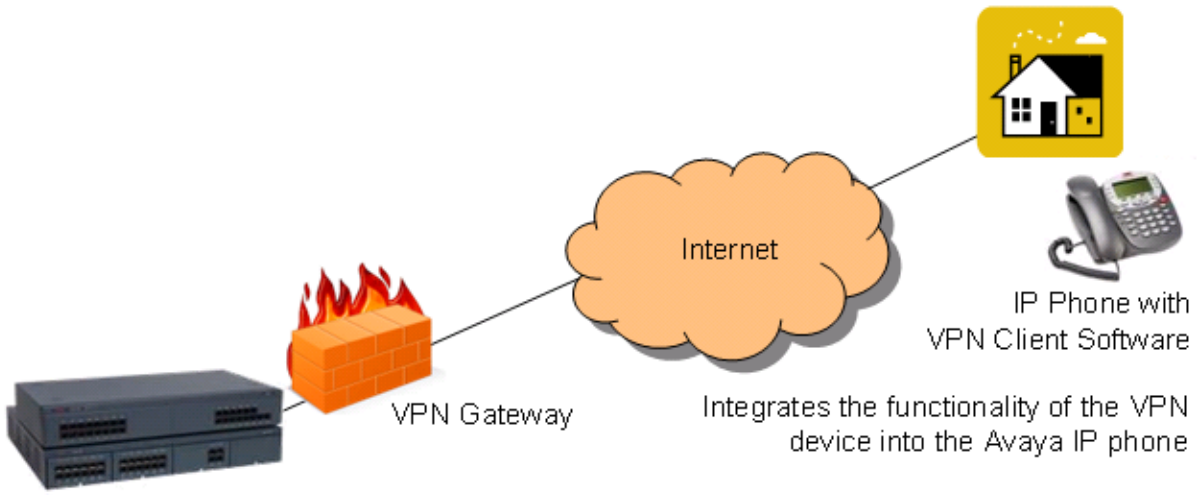
VPN Phone Software

VPN Phone is a full-featured IP Telephony solution that provides secure communication over public ISP networks to an IP Office system at the company headquarters.

It is a software-only product that runs on the standard 5610/5620/5621 or 4610/21 IP telephones. In combination with one of these phones and the most popular VPN gateway products, the software extends enterprise telephony to remote locations.

VPN Phones offer the full IP Office telephony features that are available on IP Office IP phones at the users desktop in a remote location like a home-office:

Licenses for VPN Phone are controlled by IP Office.



VPN Phone is ideal for IP Office customers supporting "work-at-home" users:

- Virtual Office workers
- Remote workers
- Remote call center
- Business continuity support
- Very small locations that require a single phone only
- Temporary installations such as conferences, off-site meetings, and trade shows

VPN Phone has been tested with a number of VPN-gateways from major vendors like Cisco or Juniper as well as with smaller VPN-access devices from companies like Adtran, Kentrox, Netgear, and SonicWall. Refer to the support pages (support.avaya.com) for a list of available application notes on VPN-gateways.

© 2008 AVAYA All rights reserved.

Other Ranges

Avaya has a wide range of communication products so we do our best to support as many telephones from other Avaya product families established in the global market such as MERLIN MAGIX and DEFINITY.

4400 Series

4406D Telephone

This range of telephones is only available in North America.

The 4406 supports the following features:

- 6 Programmable call appearance/feature keys with twin lamps.
- 8 Fixed Feature Keys: Speaker, Mute, Hold, Volume Up and Down, Conference, Transfer, Redial.
- 2 x 16 Character Display.
- Message waiting indicator.
- Two-way hands free speaker phone.
- Hearing aid compatible.
- Optional wall mounting/desk stand.
- Connects to an IP Office DS (Digital Station) port.

Note that this telephone does not support integrated directory access on the IP Office. This phone does not support personalized ringing.

This phone is not supported on the IP500 DS8 Extension Card. For the IP500 it will work on an external Digital Station Expansion Module.

© 2008 AVAYA All rights reserved.

4412D Telephone

This range of telephones is only available in North America.

The 4412 supports all of the features of the 4406 with the following differences:

- 12 Programmable call appearance/feature keys with twin lamps.
- 12 Programmable feature keys without lamps (not suitable for call appearance features).
- 4 Display Navigation Keys, right of the display: Menu, Previous (<), Next (>), and Exit.
- 4 Display Soft Keys below the Display.
- 8 Fixed Feature Keys: Speaker, Mute, Conference, Transfer, Redial, Hold, Volume Up/Down.
- DSS port to support 2 DSS4450 adjuncts; Auxiliary power required.
- 2x24 Character Display.
- Two-way hands free speaker phone.
- Optional wall mounting/desk stand.
- Connects to an IP Office DS (Digital Station) port.

Note: A maximum of twenty-seven 4412D telephones are supported on the DS30 (version 2) expansion module at PCS level 5. Earlier DS30 expansion modules will only support sixteen of these telephones.

This phone does not support personalized ringing.

This phone is not supported on the IP500 DS8 Extension Card. For the IP500 it will work on an external Digital Station Expansion Module.

© 2008 AVAYA All rights reserved.

4424D Telephone

This range of telephones is only available in North America.

The 4424D supports all of the features of the 4406 with the following differences:

- 24 Programmable call appearance/feature keys with twin lamps.
- 8 Fixed Feature Keys: Speaker, Mute, Conference, Transfer, Redial, Hold, Volume Up and Down.
- 4 Display Soft Keys below the Display.
- 4 Display Navigation Keys, right of the display: Menu, Previous (<), Next (>), and Exit.
- DSS port to support 2 DSS4450 adjuncts. Auxiliary power required.
- 2 x 24 character display.
- Connects to an IP Office DS (Digital Station) port.

Note: A maximum of twenty-four 4424D telephones are supported on the DS30 (version 2) expansion module at PCS level 5. Earlier DS30 expansion modules will only support sixteen of these telephones.

This phone does not support personalized ringing.

This phone is not supported on the IP500 DS8 Extension Card. For the IP500 it will work on an external Digital Station Expansion Module.

© 2008 AVAYA All rights reserved.

DSS4450 Unit

The DSS4450 works in association with the 4412D and 4424D telephones, each of which can support up to two DSS4450 adjuncts.

Each DSS4450 provides an additional 60 programmable keys with single red lamps except for the bottom two rows that have green lamps. The DSS4450 requires an auxiliary Avaya power supply unit and must be used with the cables supplied.

IP Office supports a maximum two 4450 units on each Digital Station expansion module, including the 406 V2 control unit.

This phone is not supported on the IP500 DS8 Extension Card. For the IP500 it will work on an external Digital Station Expansion Module.

© 2008 AVAYA All rights reserved.

Analog Telephones

As well as providing a lower cost alternative to system specific telephones, analog telephones can still deliver a high degree of functionality on IP Office. They are particularly appropriate in applications where users require lower entry costs and can be used with Phone Manager for a high proportion of call control.

Analog telephones that are compatible with caller display functionality can display the telephone number of the calling party if available. Simple programming of IP Office can convert that numeric display in to the company name associated with that number.

Feature activation by analog telephones is via short codes. IP Office is pre-programmed with a default set of short codes but these can be changed to mimic a legacy telephone system as required.

Avaya would like to stress that although most analog phones will work on IP Office - Avaya cannot guarantee that all analog phones in every region of the world will work on the IP Office.

- Analog phones connect to IP Office via ports marked PHONE ports.

© 2008 AVAYA All rights reserved.

Avaya 6200 Analog Telephone (North America)

The 6200 range of telephones are single-line analog phones that require one tip and ring pair for operation. This series of telephones have a Ringer volume control on the side of the telephone and a Handset volume control on the front of the phone. They use DTMF dialing only and support the Positive Disconnect function. In addition, these phones have a Message light, a recall button that allows access to system features, a redial button that allows automatic redial, a hold button with a single associated light, and a data jack on the rear of the telephone. The 6219 phone adds 10 programmable dialing buttons and the 6221 phone adds a built-in speakerphone with mute capability.

Analog Telephone Features	6211	6219	6221
Programmable buttons (10 buttons)	✗	✓	✓
Program Keylock	✗	✓	✓
Pause	✗	✓	✓
Redial	✓	✓	✓
Speaker	✗	✗	✓
Flash	✓	✓	✓
Hold (with indicator light)	✓	✓	✓
System Hold	✗	✓	✓
Mute	✗	✗	✓
Handset Volume Control	✓	✓	✓
Ringing Volume Control (3 position)	✓	✓	✓
Ringing Patterns (2)	✓	✗	✗
Personalized Ringing	✗	✓	✓
Message Waiting Light	✓	✓	✓
Desk/Wall Mount	✓	✓	✓
Data Jack	✓	✓	✓
Colors	White/Grey	White/Grey	White/Grey
Ringer Equivalency	0.7A, 1.6B	0.5A, 1.5B	0.5A, 1.5B
Hearing Aid Compatible	✓	✓	✓
Positive Disconnect	✓	✓	✓
DTMF Dialing	✓	✓	✓
Specialty Handset Support	✓	✓	✓

© 2008 AVAYA All rights reserved.

Interquartz Gemini Phones (EMEA and APAC)

Avaya have tested the new generation Interquartz Gemini analog telephones with IP Office to ensure that telephone and system are compatible. The Gemini phones offer good value for money without compromising on quality. Their stylish design and rugged build quality make them a popular choice for buyers on a limited budget.

For sales enquiries and product information contact Interquartz at avaya-enquiries@interquartz.co.uk.

Basic telephone 9330-AV

- Visual Message Waiting Indication.
- Locking mute button with LED indicator.
- Last number redial.
- Recall button.
- Ringer volume adjust.
- Ringer indicator light.
- Wall mountable - no additional bracket required.
- Hearing aid compatible.
- Rubber feet to minimize slippage

© 2008 AVAYA All rights reserved.

CLI Feature phone 9335-AV

All features of 9330-AV plus:

- Caller ID with 80 memories (shows date, time and new/repeat/answered/unanswered calls) .
- Large 3 line LCD display.
- IP Office feature activation through programmable keys.
- 100 name and number personal directory.
- 20 lockable direct access memories.
- Full hands-free working.
- Headset port.
- Switchable Time Break Recall 100 / 200 / 300 / 600 ms.
- Call timer.
- Alphanumeric keypad.
- Last number redial with 5 memories.

© 2008 AVAYA All rights reserved.

Hotel Phone 9281-AV

- Removable inlay card for personalized logo printing.
- Triple standard message waiting light (high voltage, reverse polarity and voltage drop).
- 10 non-volatile memories.
- Ringer indicator light.
- Ringer volume and pitch adjustment.
- Last number redial and Recall button.
- Hearing aid compatible.
- Wall mountable – no additional bracket required.
- ELR/TBR switchable.
- MF Only.

© 2008 AVAYA All rights reserved.

Doorphone Entry Systems for IP Office

Doorphones offer convenience and security. Depending on the needs of the environment, door phones may allow internal users to not only speak with someone who is outside, but also to easily allow the visitor entrance to the facility or residence. Doorphones can be connected to the Avaya IP Office base unit in a variety of ways, providing design flexibility based upon needs.

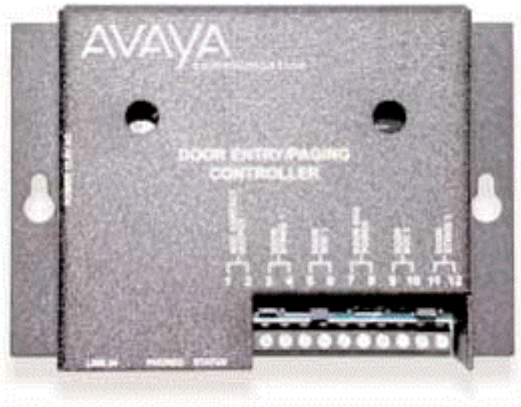
All of the IP Office base units include an external output port. Connections of doorphones to these ports enable the user to gain

access to the premises through default system short codes, through the optional Phone Manager Pro application, and through the optional VoiceMail Pro application. The flexibility of the IP Office provides the ability for short codes to be customized to a code more desirable for users. By using the Phone Manager Pro application, users can label the icons within the application a descriptive name such as Receiving Door or Front Door. The flexibility of VoiceMail Pro allows the visitor to enter a predetermined code from the phone granting access. This scenario is particularly useful in areas when co-workers are working at another site. Additionally, many doorphones can be connected to station or trunk ports available on IP Office.

The Avaya IP Office system offers three doorphone solutions to choose from:

- Avaya Universal Doorphone System (North America)
- Kalika Communications Doorphone Entry System (EMEA)
- Interquartz Doorphone (EMEA)

Avaya Universal Doorphone System:



- System consists of a controller and a speaker.
- The speaker is mounted securely on the wall and is connected to the controller, which normally resides in the equipment room. The controller is connected to a trunk port.
- Users with the trunk appearance will be notified when a visitor has pressed the Push button located on the weatherproof speaker.
- Each controller supports two speakers, for example Front Door and Back Door.
- Custom ringing mode distinguishes doorphone calls from external calls.
- Call waiting tones indicate which doorphone is calling and distinguish a doorphone call from an external line call.
- Calls can be placed on hold when visitors call from the doorphone.
- Commercial or residential security is provided via two-way hands-free communication from a door or gate.

© 2008 AVAYA All rights reserved.

Kalika Communications Doorphone Entry System:

- Supports a doorway intercom system with up to 46 buttons.
- The system can be programmed to enable multiple extensions to answer and control the operation of the door and can be used with both single and multiple door entry systems.
- It is ideal for apartment complexes or where different companies occupy different floors and require their own unique door entry solution.
- The Kalika Communications Control Unit is available in several versions and is equipped to provide two-way voice communications, electrical lock control and label lamps.
- It is weatherproof and remotely programmable.
- For sales enquiries and product information contact Kalika Communications at info@kalika.co.uk.

Interquartz Doorphone:

- Choice of models (1, 2, and 4 button).
- Slim design (16mm thick).
- Strong aluminium casing.
- Optional PC configuration.
- Remotely programmable via DTMF.
- Connection via analog extension port or trunk port.
- Relay lock control.
- Backlit inlay cards.
- Internal heating system.
- Day/Night service.
- Combination lock control.
- For sales enquiries and product information contact Interquartz at support@analogphonesavaya.com.

© 2008 AVAYA All rights reserved.

Headsets

Avaya offers ergonomically designed communication headsets and amplifiers for the Avaya IP Office telephones. This full line of professional and contact center solutions set the standard in sound quality and durability. Avaya headsets are designed for maximum, all-day comfort and are available in styles that suit nearly any wearer and any usage pattern.

Whether you want the freedom to communicate hands-free while working at your desk, or the ability to roam while talking, you will find a solution that suits your individual needs.

To view the full range of Avaya headsets:

1. Go to http://www.avaya.com/gcm/master-usa/en-us/headsets/index.htm.
2. Identify the IP Office telephone you are using.
3. Choose an amplifier based on compatibility and features.
4. Choose the style of headset that best suits your needs. For instance, noise-canceling headsets are great in a busy office or when using VoIP telephones.

© 2008 AVAYA All rights reserved.

Summary

All Avaya telephones are designed to ensure that features and functions are easily accessible to the user - ensuring that, through ease of use, the full benefits of the system are delivered to the desktop.

The telephones listed below are the preferred and premier range of telephones for use on the IP Office. These telephones are sold worldwide in every country that the IP Office is available. This telephone range consists of both digital and IP telephones.

IP Office worldwide digital phones:	IP Office worldwide IP phones:
• 5402 Telephone 56	• 5602SW IP Telephone 56
• 5410 Telephone 58	• 5610SW IP Telephone 58
• 5420 Telephone 60	• 5621 IP Telephone 60
	• 1703 IP Telephone 47
	• 1708 IP Telephone 47
	• 1716 IP Telephone 47

In addition to the telephones above, the IP Office supports a wide range of phones as listed below. However, note that some of these phones are only available in certain countries and regions.

• 4406D Telephone 85	• 3616 Executive Wireless (WiFi) Phone 75
• 4412D Telephone 86	• 3620 Healthcare Wireless (WiFi) Phone 76
• 4424D Telephone 87	• 3626 Ruggedized Wireless (WiFi) Phone 77
• 4450 DSS Unit 88	• 3641 Ruggedized Wireless (WiFi) Phone 78
• T3 Compact (Upn and IP) 64	• 3645 Ruggedized Wireless (WiFi) Phone 79
• T3 Comfort (Upn and IP) 66	• 3701 IP DECT Wireless Handset 80
• T3 Classic (Upn and IP) 65	• 3711 IP DECT Wireless Handset 81
• Interquartz Gemini 9281-AV, 9330-AV and 9335-AV analog telephones 91	• 3810 Wireless Telephone 82

Phones supported worldwide in addition to 5400 Series:	IP phones supported worldwide in addition to the current 5600 and 1700 Series:
• 2402 Telephone 56	• 4601 IP Telephone 55
• 2410 Telephone 58	• 4602SW IP Telephone 56
• 2420 Telephone 60	• 4610SW IP Telephone 58
• XM24 DSS Unit.	• 4621SW IP Telephone 60
• EU24/EU24BL DSS Unit 62	• 4625SW IP Telephone 60
• Analog Telephones 89	

Phones supported on IP Office 4.2 but no longer available for new sales:	
• 6408D Telephone	• 5601 IP Telephone
• 6416D Telephone	• 5620 IP Telephone
• 6424D Telephone	• 4620 IP Telephone
• 20DT DECT Telephone (with IP DECT only)	

- For maximum cabling distances please refer to the IP Office Installation Manual.
- Those phones that support hands free operation are intended for individual use only, not for group and conference room operation.

© 2008 AVAYA All rights reserved.

Chapter 4. Features

© 2008 AVAYA All rights reserved.

4. Features

IP Office provides a comprehensive telephony feature set to enable a fast and efficient response to a telephone call. Features such as Caller ID display and call tagging allow employees to see who is calling and who they are calling before they pick the call up. Client information can even be 'popped-up' on the user's PC.

For those who are not tied to a desk, Wireless handsets and twinning offer mobility around the office. For those out of the office, be it on the road or working from home, comprehensive and easy to use call forwarding facilities, PC Softphone and a remote access service allow them to remain in telephone contact and access centralized resources at all times.

Incoming calls can be efficiently handled using either Direct Dialling (DDI/DID) or dedicated operators. For out of hours calls or times when you just can't take calls, IP Office provides voicemail and optional Auto-Attendant services.

© 2008 AVAYA All rights reserved.

Basic Call Handling

Tones

IP Office generates the correct user tones for the geography. These tones are generated for all IP Office extension types, analog, digital and IP.

Supported tones are:

- Dial, both primary and secondary depending on geography
- Busy
- Unobtainable
- Re-order
- Conferencing tone depending on geography

Caller ID

Feature

- Display of the caller's number on incoming calls, where supplied by the service provider.
- Sending of calling number on outgoing external calls.

Benefit

- Confirmation and recognition of who is calling.
- Storage of Caller ID numbers for return calls.
- Directory name matching to Caller ID numbers.
- Screen-Popping customer records in compatible applications.

Description

Where supplied by the service provider, the IP Office can receive and use the callers Caller ID. The Caller ID is passed through to the answering phone or application and is included in any call log or history supported by the phone or application. If the Caller ID matches a number in the IP Office's Directory, the matching directory name is shown instead of the number.

Where IP Office Phone Manager, or the TAPI service is used to link to database software on the users PC, it is possible to have an automatic query performed on the supplied Caller ID and have the caller's record in front of the user before the call is answered.

For outgoing calls the IP Office can insert a system wide Caller ID or set a flag to have Caller ID withheld. For users with a direct dial number routed to their extension, that direct dial number is also used as their Caller ID for outgoing calls. Alternatively short codes can be used to specify the Caller ID that should be sent with outgoing calls.

Note that the sending and receiving of Caller ID is subject to the service provider supporting that service. The service provider may also restrict which numbers can be used for outgoing Caller ID.

Hold

A call may be placed on hold with optional Hold music. A held call cannot be forgotten as it is presented back to the extension after a timeout set by the system's administrator.

See also **Park** 101.

Toggle Calls

Toggle Calls cycles round each call that the user has On Hold to their extension locally within the system, presenting them one at a time to the user

© 2008 AVAYA All rights reserved.

Hold Call Waiting

Hold Call Waiting is a compound feature combining hold and answer and provides a convenient way to hold an existing call and answer a waiting call through a single button press.

Hold Music (Music on Hold)

The IP Office system supports up to 4 sources of music on hold - one system source which may be external, internal (WAV) or tone, plus up to 3 additional internal sources. Small Office Edition supports either one internal or one external music on hold source at a time. The internal sources are .WAV files saved either in volatile memory, or on the optional memory card in an IP500, IP406 V2 or Small Office Edition. The .WAV file must be 16bit PCM mono and sampled at 8Khz with a maximum duration of 30 seconds.

From release 4.2, the alternate sources for music on hold are selectable for use by Incoming Call Routes or Hunt Groups (not available on the Small Office Edition). On IP500 systems, each source can be up to 90 seconds long (30 seconds on IP406 V2 and IP412).

External music on hold sources connect to the 3.5mm Audio socket on all IP Office control units.

Park

As an alternative to placing a call on hold, a call can be parked on the system to be picked by another user.

The call park facility is available through the user's telephone, Phone Manager or SoftConsole. Calls are Parked against a 'park slot number' which can be announced over a paging system so the person the call is for can go to any phone and collect the call by dialling the park slot number.

For convenience Phone Manager has 4 pre-defined park buttons. On digital phones with DSS/BLF keys it is possible to program Park keys that will indicate when there is a call in a particular park slot and allow calls to be parked or retrieved.

There is a system configurable timeout that determines how long a call may remain parked before it is re-presented to the extension that originally parked the call.

Automatic Callback

Feature

- When calling an extension that is busy, set the system to call you when the extension becomes free. This feature is also called "Ringback When Free".
- When calling an extension that just rings, set the system to call you when the extension is next used. This feature is also called "Ringback When Next Used".

Benefit

- Carry on with other work and let the system initiate a call for you when the extension becomes available.

Description

Depending on the type of phone a user has, call back when free is accessed by dialing a short code while listening to internal busy tone, selecting an option from an interactive menu or pressing a programmed DSS/ BLF key. Callback when free can also be activated from Phone Manager.

You can also set a callback when free or a callback when next used using a short code without attempting a call.

Note that a user can only have one automatic callback set at any one time.

This feature is supported across the IP Office Small Community Network.

© 2008 AVAYA All rights reserved.

Direct Inward Dialing (DID/DDI)

This relies on the local telephone exchange passing all or part of the dialed number to the IP Office. This number can then be used by IP Office call routing software to route the call to an individual phone, or groups of phones. This service is typically used to reduce the workload on a reception position by giving members of staff or departments individual numbers so they can be called directly. For convenience it is common to have the extension or group number the same as the digits supplied from the network, but IP Office can convert the number to what ever number is needed by the business, within limits

In North America, T1 circuits are required for DID.

Transfer

Call Transfer allows users to transfer a call in progress to another phone number – either internal extension or external public number. The caller is placed on hold while the transfer is performed.

If the phone is put down before the destination has answered, the original caller will be automatically transferred. This is called an Unsupervised or Blind Transfer. Alternatively, a user can wait for the destination to be answered and announce the transfer before hanging up to complete the transfer. This is called a Supervised Transfer.

Unless restricted by the system administrator, the IP Office makes no differentiation between internal or external call transfers.

Distinctive and Personalized Ringing

The IP Office uses different ringing sequences to indicate the type of call, for example whether internal or external. This feature is called 'distinctive ringing'. For analog phones the distinctive ringing sequences used are adjustable. For digital and IP phones the distinctive ringing sequences are fixed as follows;

- Internal Call: Repeated single-ring.
- External Call: Repeated double-ring.
- Ringback Call: Single ring followed by two short rings.

This ring is used for calls returning from park, hold or transfer. It is also used for call back when free and voicemail ringback calls.

This feature is supported across the IP Office Small Community Network

Personalized Ringing

In IP Office the term personalized ringing is used to refer to changing the sound or tone of a phone's ring. On many Avaya digital phones, the ringer sound can be personalized. Changing the ringer sound does not alter the ring sequence used for distinctive ringing. This feature is local to the telephone and not supported on all types of telephones.

Message Waiting Indication

Message waiting indication (MWI) is a method IP Office uses to set a lamp or other indication on compatible telephones when a new message has been left for the user, either in a personal voice mailbox or in a group mailbox or call back message. When the message has been played or acknowledged, the lamp is turned off.

All Avaya digital and IP phones all have in-built message waiting lamps, and the IP Office Phone Manager application provides message waiting indication on screen

For analog phones, from IP Office 3.1 a variety of analog message waiting indication (MWI) methods are provided. Those methods are 51V Stepped, 81V, 101V and Line Reversal. The MWI method must be selected from the IP Office Manager application when configuring a system to match the properties of the analog phones. Note that the 101V signaling is only available on IP500 Phone cards and modules plus version 2 IP400 Phone 8, 16 and 30 modules, not on the IP406 system unit.

© 2008 AVAYA All rights reserved.

Visual Voice

Feature

- Provides interface to voicemail through handset display and buttons e.g. Listen, Save, Delete, Fast Forward.

Benefit

- Quick access to voicemails and commonly used messaging features.

Description

With IP Office R4.0, you can now access and control voice messages via the display on Digital or IP phones. Visual Voice requires VoiceMail Pro or Embedded Messaging, and can be used with large display LCD sets only (1708, 1716, 2410, 2420, 5410, 5420, 4610, 4620, 4621, 4625, 5610, 5620, and 5621 phones plus all T3 phones are supported).

From release 4.2, the fixed message button on certain phones can be configured to invoke visual voice. This configurable option frees the button programming required currently for Visual Voice.

Features supported are:

- access new/old/saved messages for personal and hunt group mailboxes.
- next and previous message.
- fast forward and rewind.
- pause message.
- save, delete and copy message to other users of the system.
- change default greeting.
- change password.
- change email settings (VoiceMail Pro only).

Note: Visual Voice is NOT available on VoiceMail Lite.

© 2008 AVAYA All rights reserved.

Advanced Call Handling

Description
In larger businesses or businesses with greater reliance on the telephone for internal and external communications some of the more advanced features will improve efficiency and customer service. Features like Pick-Up which permit users to take a call for a colleague who is temporarily away from their desk, of Absence Text which can quickly give information to internal callers about a person's availability.

Absence Text

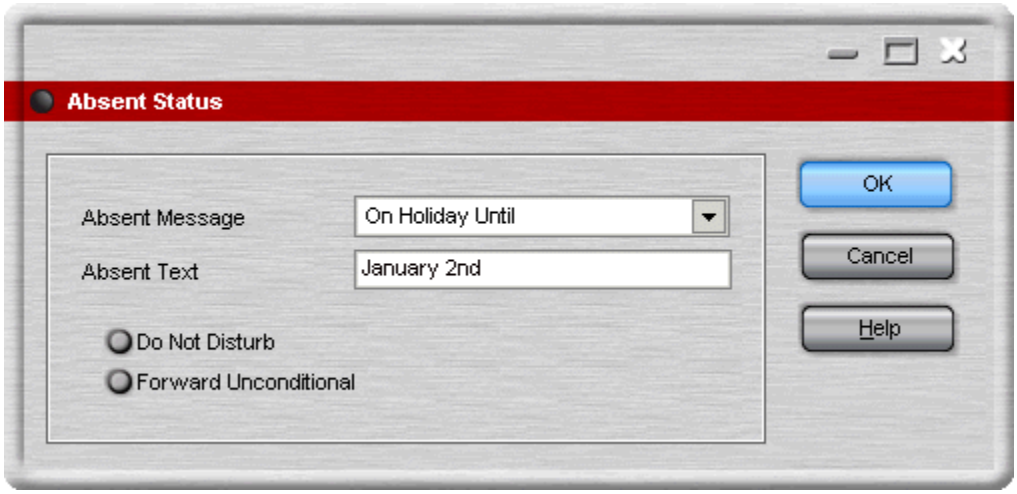
Feature

- Display a text message on the user's phone and IP Office Phone Manager application.
- Display the same message on other internal phones and IP Office applications when calling the user.

Benefit

- Inform other internal users of your current status and likely availability.

Description
Any user can set Absence Text on their phone, even users of standard analog phones, but it can only be displayed on selected display phones, Phone Manager and SoftConsole that call the user. Most supported feature phones give the option of adding some text, for example, "At lunch until 16:00".



When a user has an absence text message set, call processing is not affected to the user and they still have the choice of using features like Do Not Disturb or Forward on No Answer as appropriate. Phones that support the interactive setting of Absence Text will also display it on the users own phone for the benefit of people who come to their desk. There are 10 predefined strings for Absence Text:

- None (no text message)
- "On vacation until"
- "Will be back"
- "At lunch until"
- "Meeting until"
- "Please call"
- "Don't disturb until"
- "With visitors until"
- "With cust. til"
- "Back soon"
- "Back tomorrow"
- Custom

All may have additional text entered, eg message 4 plus 10:00 will show "Meeting until 10:00" and the text strings are localized to the system language

This feature is supported across the IP Office Small Community Network

© 2008 AVAYA All rights reserved.

Call Tagging

Feature

- Display a text message on the user's phone, or Phone Manager application, when a call is presented to it.

Benefit

- Provide additional information about the call.

Description

This feature is used to provide additional information about the call to the targeted user before they answer it. Call Tagging may be used when transferring a call from Phone Manager or Soft Console to give caller info if the user doing the transfer is not able to announce the call.

It is possible to add a tag to a call automatically using CTI and IP Office VoiceMail Pro. This is also possible based on an Incoming Call Route. On some telephones, displaying the Tag may mean that it is not possible to display the usual call source and target information.

Reclaim Call

Feature

- The ability to recover, or reclaim, the last call that was at your phone but is now ringing or is connected elsewhere.

Benefit

- If you just miss a call and it goes to voicemail or call coverage, you can get the call back while it is still being presented or connected through IP Office

Description

This is a special version of the Acquire Call feature that only applies to the last call at your extension.

Hunt Group Enable/Disable

Feature

- The ability for a user to enable or suspend their membership of Hunt Groups.

Benefit

- A user may need to temporarily join or leave individual hunt groups, for example to cover a peak of calls without changing the system programming.

Description

A team supervisor or administrator may not usually take calls for a team but at times of high traffic they may join the group to take calls and when the peak is over leave the group to resume their regular tasks. To use this feature the User must be configured as a member of the Hunt Group by the systems administrator, it is not possible for a User to arbitrarily join a Hunt Group that they have not been identified as a member of.

© 2008 AVAYA All rights reserved.

Call Waiting

A User may not want people calling them to receive busy tone if they are already on another call, but have the call receive ring tone and have some kind of alert that there is a call waiting. The user can then decide to finish or hold the current call and answer the one that is waiting. The amount of information that is available about the call that is waiting depends on the type of phone the user has, or if they are using Phone Manager.

As Call waiting tone can be disruptive it is possible to turn the feature on or off and even suspend it for a single call – useful for conference calls.

Do Not Disturb (DND)

This is the ability to temporarily stop incoming calls ringing at a user's telephone. It will prevent the user from receiving Hunt Group calls and give direct callers either voicemail (if enabled) or a busy signal. This feature can be enabled/disabled from the phone or via the Phone Manager application.

It is possible to have some calls bypass the DND setting and ring the phone. For example a manager might have their secretary's extension number on the DND exceptions list. The exceptions list can be easily managed by the Phone Manager application. Both internal and external numbers can be on the exception list.

Dial Plan

IP Office has a very flexible numbering scheme for extensions, hunt groups and feature commands. While the system has default numbering for feature codes and extensions, they can all be re-defined. Default extensions and hunt groups have 3 digit numbers starting at 200 but these can be changed from 2 to 9 digits through the IP Office Manager application. There is a default set of feature access "short codes," but these can be changed to what ever the end user requires, within limits. This is useful for example, if IP Office is replacing a system where DND was accessed by dialling *21, it is possible to change the IP Office Short Code to mimic the code of the replaced system.

In certain countries IP Office can support a Secondary Dial Tone when an access digit is dialled, though this limits some functionality like Alternate Route Selection (ARS). IP Office can also be configured to work without line access digits, by analyzing digits as they are dialled and determining if they are for an internal number or should be sent out on a line – this is valuable in SOHO installations where users will not necessarily be used to dialling an access digit for an outside line.

Paging

All Avaya digital and IP phones supported on the IP Office that have loudspeakers can be used to receive broadcast audio messages without having to install a separate paging system. Paging can be to individual phones or groups of phones.

Analog extension ports can be configured for connection to external overhead paging systems, usually through an adapter, such that a port can be included in a paging group to permit mixed phone and overhead paging.

Some Avaya digital and IP phones are able to answer a page by pressing a key while the page is going on, this terminates the page and turns it into a normal call.

This feature is supported across the IP Office Small Community Network

Intrude

The Call Intrude feature allows a user, if permission through IP Office Manager is given, to join an existing conversation whether this is an internal or external call.

A user with the "Can Intrude" option can join a call on any extension on the system, however, a User with "Cannot be Intruded" setting would prevent others from joining their call.

© 2008 AVAYA All rights reserved.

Inclusion

This feature enables selected users to intrude on calls that are already in progress. The intruding party intrudes on the existing call and all parties hear a tone. The speech path is enabled between the intruding party and the called user, the other party is forced onto hold and will not hear the conversation. On completion of the intrusion the called party speech path is reconnected to the original connected party. The feature is enabled or disabled on a per user basis through the Manager application.

Private Call

Users can set a status of private call using short codes or a programmed button. Private calls cannot be recorded, intruded on, bridged into or monitored.

Hot Desking

Hot Desking allows a number of users non-exclusive use the same extension. Each user logs in with their own identity so they can receive calls and can access their own Voicemail and other facilities. For example, sales personnel who visit the office infrequently can be provided with telephony and Voicemail services without being permanently assigned a physical extension. When finished, they simply log off to make the extension available to others or if users log on at another phone, they are automatically logged off the original extension.

Remote Hot Desking

Feature

- The ability for a user to Hot Desk to other locations within the Small Community Network.
- Available on Digital, Analog and IP phones.

Benefit

- A user can make and receive calls from any office as if using the phone on their own desk.
- Single number, improved mobility and easy access to familiar features.
- Great for consultants, managers, lawyers working on different offices on different days.

Description

IP Office 4.0 supports remote hot desking between IP Office systems within a Small Community Network. The system on which the user configured is termed their 'home' IP Office, all other systems are 'remote' IP Offices. To log on at a remote IP Office requires that IP Office to have a Small Community Advanced Networking license. A license is not necessary on the user's home IP Office.

- **User Settings**

  When a user logs on to a remote IP Office system, all their user settings are transferred to that system.

- The user’s incoming calls are rerouted across the SCN.
- The user’s outgoing calls use the settings of the remote IP Office.
- However some settings may become unusable or may operate differently. For example if the user uses a time profile for some features, those feature will only work if a time profile of the same name also exists on the remote IP Office.
- **Break Out Dialing**

  In some scenarios a hot desking user logged on at a remote system will want to dial a number using the system short codes of another system. This can be done using either short codes with the Break Out feature or a programmable button set to Break Out. This feature can be used by any user within the Small Community Advanced Network but is of significant use to remote hot deskers.

Note: Remote Hot Desking is not supported for use with CCC. Features handled by the telephone itself are not affected by Hot Desking (e.g. call log and phone speed dials).

© 2008 AVAYA All rights reserved.

Relay On/Off/Pulse

IP Office is fitted with two independent switch outputs for controlling external equipment such as door entry systems. Control of these switches is via allotted handsets allowing the switches to be opened, closed or pulsed as required. Control of switches is also accessible via Phone Manager Pro, SoftConsole and VoiceMail Pro.

Pickup

Call Pickup allows a user to answer a call presented to another extension. Types of call pickup include:

- Pick up any call ringing on another extension.
- Pick up a Hunt Group call ringing on another extension, where the user must be a member of that Hunt Group.
- Pick up a ringing call at a specified Extension.
- Pick up any call ringing on another extension that is a member of the Hunt group specified.

This feature is supported across the IP Office Small Community Network

Call Recording

Where IP Office has VoiceMail Pro installed it is possible to record a call and save the recording to the user's mailbox, a group mailbox or the voice recording library. For example, this is useful when a caller is going to give detailed information like an address or phone number and the caller will hear a warning message or tone that the call is being recorded in some countries. Where call recording is required for Quality Assurance, it is possible to set the IP Office system to automatically record a percentage of calls for later review.

Beginning with IP Office R4.0, any call (normal, conference, or intrusion) and any phone type (including IP) can be recorded. Where "advice of recording" needs to be played, IP Office will ignore Voicemail port licensing if an insufficient number of voicemail channels have been licensed.

Note: for IP phones, a VCM channel will be required for the duration of the recording.

Telecommuter Mode

Phone Manager Pro allows the making and receiving of calls and the retrieving of voicemails from an external phone number as if they were in the office, with Phone Manager providing the call control. The typical scenario is the remote worker that occasionally works from home or from a hotel room.

This feature also provides billing convenience and potential cost savings for remote workers and mobile work force as all the calls are established by IP Office: there is no need to check bills, nor to pay for expensive hotel calls.

© 2008 AVAYA All rights reserved.

Twinning and Mobility

Twinning allows a primary extension and a secondary number (extension or external) to operate together as a single telephone, When a call is presented to the primary phone the secondary will ring. If the primary telephone does not ring, for example in Do Not Disturb, the secondary phone will not ring.

Internal Twinning

This is typically used in scenarios like workshops or warehouses where team supervisors may have a desk with a fixed phone but also have a wireless extension (e.g. DECT). When a call is made from either twinned phone, the call will appear to have come from the primary phone (when the secondary is an extension on the IP Office system). Other users of the system need not know that the supervisor has two different phones. The supervisor's Coverage Timer and No Answer Time are started for the call and if the call is not answered within that time, the call will be delivered to available coverage buttons (if applicable) and then Voicemail (if applicable).

Mobile (external) Twinning

A typical use for Mobile Twinning would be for a salesperson who has a desk extension but is often out of the office. They can give out one phone number to clients, but it will ring their mobile/cell phone as well as the desk extension. External mobile devices can be set as twinning targets even if the primary telephone is logged out (e.g. in the case of a "virtual" extension).

From IP Office 4.2 onwards, the following additional Mobility features are available in addition to simple mobile twinning (but need to be separately enabled per user in IP Office Manager):

Mobile Call Control

This allows a user who has answered a twinned call on their mobile phone to dial a DTMF sequence **(star-star) to put the call on hold and receive IP Office system dial tone. The user will then be able to perform supervised and un-supervised transfers, shuttle (switch between held calls), and conference (Conference add and transfer to a meet-me conference).

Avaya one-X™ Mobile Client Support

IP Office 4.2 supports the Avaya one-X Mobile "Single Mode" Client running on selected Symbian Single-Mode or Windows Mobile 5 or 6 handsets. This provides a graphical interface for call control and allows a one-number service for both incoming and outgoing calls. For a full list of supported handsets see http://support.avaya.com. IP Office systems supporting one-X Mobile require a dedicated DID/DDI number for each feature required (e.g. hold, transfer).

Mobile Call Control and one-X Mobile support require twinned calls to be routed through specific trunk types on IP500 only:

- IP500 PRI-U (single or dual) daughter cards
- IP500 Dual or Quad BRI daughter cards
- SIP trunks (RFC 2833)

In IP Office 4.2 the existing Mobile Twinning license has been renamed to Mobility license. The license now enables the use of Mobile Call Control and one-X Mobile client features as well as Mobile Twinning. An instance of the license will be consumed for every user who has Mobility Features checked in IP Office Manager.

With either internal and external twinning, users may be allowed to enter a twinned number, or may just be able to activate/deactivate the twinning function depending on administrative settings. Such user activation can be achieved through deskphone buttons, Phone Manager or VoiceMail Pro.

© 2008 AVAYA All rights reserved.

The following types of calls are eligible for twinning:

	Internal twinning	External (mobile) twinning
Any internal call on a Call Appearance button	✓	✓
Internal or external calls transferred to the extension	✓	✓
Direct Dial calls to that extension	✓	✓
Hunt Group Calls	✓	✓
Calls forwarded from another extension	✓	✓
Line Appearance calls (configurable)	✓	✗
Bridged Appearance calls (configurable)	✓	✗
Coverage calls (configurable)	✓	✗
Automatic Intercom calls	✓	✗
Returning transferred, held or park calls	✓	✓*
Callback calls from the system (Transfer and Park Return)	✓	✗
Paging Calls	✓	✗
Follow Me calls	✓	✗

*Requires IP500 BRI, PRI or SIP trunks and Mobile Call Control to be enabled for that user.

© 2008 AVAYA All rights reserved.

Key and Lamp Operation

IP Office offers a full range of Key and Lamp features on Avaya feature phones. These features include; Line Appearance, Call Appearance, Bridged Appearance and Call Coverage. As these features require a phone with buttons and indicators, the features are only supported on certain Avaya digital and IP phones. Key and Lamp operation is not supported on analog phones.

IP Office can have a ring delay set on each appearance button to allow time for the target number to answer before other extensions ring, or visual alert only without ring.

In Key and Lamp operation, IP Office supports up to 10 buttons on each telephone and 10 telephones with the same line appearance.

Appearance Buttons

Feature

- Use the programmable buttons available on Avaya digital and IP telephones to represent individual calls.
- Answer, originate and join calls by pressing the appropriate appearance buttons.

Benefits

- Indication of calls connected and calls waiting.
- Handling of multiple calls from a single phone.

Description

Many Avaya digital and IP telephones supported by IP Office have programmable buttons. These buttons can be assigned to appearance functions that allow the handling of calls. These functions are:

- Line Appearance Buttons

  Used to indicate make and answer calls on a specific external trunk.

- Call Appearance Buttons

  Used to handle multiple incoming and outgoing calls from a user's extension.

- Bridged Appearance Buttons

  Used to match the call appearance buttons on a colleagues extension.

- Call Coverage Buttons

  Used to indicate unanswered calls ringing at a colleagues extension.

Line Appearance

A Line Appearance is a representation of a trunk line on the IP Office system where the indicator tracks the activity on the Line. Only external calls can be answered or made on Line Appearances. Line appearances can be used with Analog, E1 PRI, T1 PRI and BRI trunks PSTN trunks. They cannot be used with E1R2, QSIG and IP trunks.

© 2008 AVAYA All rights reserved.

Call Appearance Buttons

Feature

- Uses a programmable button on the Avaya digital and IP telephone to represent an incoming or outgoing call.
- Separate buttons are used to represent each simultaneous call that the user can make or answer.
- Where possible, the status of the calls (ringing, connected or held) is indicated by the button indicator.

Benefit

- Call appearances allow a single user to make, answer and switch between multiple calls by pressing the appropriate call appearance button for each call.

Description

On Avaya IP Office digital and IP telephones that have programmable buttons, those buttons can be set as call appearance buttons through the IP Office Manager application. The number of call appearance buttons set for a user determines the number of simultaneous calls they can make and answer.

Note that the use of call appearance buttons overrides IP Office call waiting features. It is only when all call appearances are in use that subsequent callers receive either busy tone, voicemail or follow a forward on busy action

When call appearance buttons are used, a minimum of three call appearance buttons is recommended where possible, although some phones are restricted to two call appearance buttons by the number or design of their programmable buttons.

Bridged Appearance Buttons

Feature

- Allow the user to have an appearance button that matches another user's call appearance button.

Benefit

- Answer and make calls on behalf of the other user.
- Audible indication of calls presented to the bridged user, where programmed
- Visual indication of when the other user has calls presented, held or connected.
- Join and exchange calls using the paired call appearance and bridged appearance buttons.

Description

A bridged appearance button matches the activity on one of another user's call appearance button. For example, when the call appearance shows a ringing call, the bridged appearance button will also show the ringing call and can be used to answer that call.

Similarly, if the bridged appearance button is used to make a call, the call activity is shown on the matching call appearance button. The call appearance button user can join or takeover the call using their call appearance button.

Bridged appearance buttons allow paired 'manager/secretary' style operation between two users, and are only supported for users who have call appearance buttons.

Call Coverage

Feature

- Allow unanswered calls to alert at other user extensions and be answered there before being forwarded or going to voicemail.

Benefit

- Provide users the opportunity to answer colleague’s unanswered calls before they go to voicemail.

Description

When a user has an unanswered call ringing, after a configurable delay, the call will also start alerting on any call coverage buttons associated with the user on other extensions. The call can then be answered by pressing the call coverage button. If still unanswered the call is forward or goes to voicemail as normal.

The time a call rings before also alerting on any associated call coverage buttons can be adjusted for each user.

© 2008 AVAYA All rights reserved.

Outbound Call Handling Features

Every business needs to make calls, but depending on the type of business these calls may need to be treated in a special way, such as recorded against a project or client through the use of Account Codes. A business may have several sites linked via a private network but certain users, like customer services agents, may need to be able to call colleagues in other offices even when the network is busy, while other users can wait for a line to come free, Least Cost Routes can automatically translate the internal number to a direct dial call over the public network while other users wait.

Account Codes

Feature

- Associate an account code with a call.
- Validate account codes used against list stored by the IP Office.
- Include the account code used with call log details.

Benefit

- Through the call records, group calls by account code for the purpose of call costing and tracking.
- Restrict outgoing calls by requiring users to enter a valid account code.

Description

IP Office stores a list of valid account code numbers. When making a call or during the call, the user can enter the account code they want associated with that call. IP Office will check the account code against its list of valid codes and request the user to re-enter the code if it is not valid. For incoming calls, the Caller ID can be used to match it with an account code from the IP Office's list of valid codes and report the account code with the call for billing.

Individual users can be set to Forced Account Code operation where they are required to enter a valid account code before making external calls. By using IP Office Short Codes it is possible to identify certain numbers or call types as requiring a valid account code before permitting the call to proceed, for example long distance or international numbers. Analog phone users can only enter account codes before making a call or in response to an audible system prompt to enter a code when making the call.

Account codes can also be entered through the IP Office Phone Manager application, a system wide setting, determines whether Phone Manager will display a list of account codes from which users can select the code required or will hide the account code list.

In all the cases above, the account code entered is included with the call details in the IP Office's call record output. (CDR and SMDR).

Authorization Codes

Authorization codes allow an IP Office user to go to another extension on the system and make calls using their personal toll restrictions; this may grant the user greater or fewer privileges than the normal owner of the extension they use. Since Authorization Codes are independent of Account Codes, the user has to enter both if the required by the system configuration. All entered codes are logged in CDRs.

Dial Emergency

Dial emergency is an IP Office Short Code and, permits certain numbers to be dialed regardless of call barring or a phone being logged off.

© 2008 AVAYA All rights reserved.

Call Barring

Feature

- It is possible to prevent or allow calls to certain numbers such as international numbers or premium rate numbers for individual users or on a system wide basis.

Benefit

- Restrict the dialing of specific numbers or types of numbers system wide.
- Restrict certain users from dialing specific numbers or types of numbers.

Description

IP Office supports call barring at many levels. Short codes can be used at the system or individual user level to block the external routing of specific numbers or types of numbers. Typically the barring short codes are set to return busy tone, however they could route the call to an alternate number or to a Voicemail service that returns a 'barred dialing message'.

For users, the short codes can be allocated to a User Rights template. This template is then applied to the Users whose calls need restriction. In addition to barring the dialling of certain numbers, IP Office can be set to bar the forwarding of calls to external numbers on a per user basis.

Alternate Route Selection (ARS)

IP Office supports Alternate Route Selection, which is more flexible and easier to configure than Least Cost Routing (LCR). If a primary trunk is unavailable, then ARS provides automatic fallback to an available trunk (e.g., analog trunk fallback if a T1 or SIP trunk fails, or use PSTN for SCN fallback).

By configuring ARS, calls may be routed via the optimum carrier. Time profiles can also be used to allow customers to take advantage of cheaper rates or better quality at specific times of day.

Multiple carriers are supported. For example, local calls are to go through one carrier between specific hours and international calls through an alternative carrier. Carrier selection using 2-stage call set up via in-band DTMF is possible. It is possible to assign specific routes on a per user basis, e.g. only allow expensive routes to be used by critical staff.

Note: Existing LCR configurations are automatically converted to ARS when upgrading to 4.0

Maximum Call Length

This feature allows the system to control the maximum duration of any call based on the dialed number. This could be used for controlling calls to cellular networks or data calls made over the public network to ISPs.

PIN Restricted Calling

See **Account Codes** 113.

© 2008 AVAYA All rights reserved.

Forwarding

This is the ability to forward a user's calls to another extension or external number such as a Mobile/Cell Phone. Calls can be forwarded in a number of ways and if the call is not answered at the forward destination it will go to IP Office voicemail if enabled for the user and call supervision is available. There are three separate forward destinations, one for forwarding on busy one for no answer and one for forward unconditional. Once the numbers have been entered, the user can toggle the forwarding to be active or not as required without having to re-enter the numbers.

If the user is a member of a hunt group, some types of Hunt Group calls can also follow forward unconditional. Users can select if forwarding is applied to external calls only, or all calls. Call forwarding is processed after Do Not Disturb and Follow-Me conditions are tested.

Associated Features

- Do Not Disturb (DND)
- Voice Mail (VM)
- Follow Me
- Hunt Groups
- No Answer Interval

Precedence

- Forward Unconditional
- Forward Busy
- Forward No Answer

Forward on Busy

If enabled, this forward will be triggered when the user is busy and another call is routed to them, but does not include calls for a hunt group that they may be a member of. A user is normally considered to be busy when they are on a call but depending on call waiting settings and key and lamp features this may not be the case.

Forward on No Answer

This forward is triggered if a call has been ringing for a user but they haven’t answered it within the configured answer time, this includes calls that have been indicating call waiting if enabled.

Forward Unconditional

This sends all calls for the user to the forward unconditional number, but if the call is not answered within a user’s timeout period the call will be sent to IP Office voicemail, if enabled.

Forward Hunt Group

Calls for a hunt group that the user belongs to can also follow forward unconditional. The hunt group must be set for either hunt or rotary ring type and if the call is not answered at the forward destination it will follow the hunt group call handling instead of going to voicemail. This can be particularly useful in a sales or support environments where a number of people may be out of the office on Mobile/Cell Phones and still participate in the hunt group as if in the office.

© 2008 AVAYA All rights reserved.

Follow Me

Follow-Me is similar to Forwarding except that the destination can only be an extension on the same IP Office as the user making use of the feature. Follow-Me is typically used when a user is going to be working away from their desk, for example in a workshop. All the call settings the user has on their main phone will apply to calls that follow the follow-me feature, including forward on busy or no answer.

Follow-Me can be set either from the users main phone – Follow-Me To – or from the phone where they want calls to be received – Follow-Me Here. Several people can have their phones forwarded to a follow-me destination and if the phone has a display it will indicate who the call is for.

© 2008 AVAYA All rights reserved.

Avaya Digital and IP Phones

Programmable Buttons

As well as the usual dialing keys, Avaya digital and IP phones have dedicated function buttons like Mute, Volume, Hold, Conference and Transfer. In addition to these, on many Avaya digital and IP phones there are keys that can be programmed with a range of selected special functions. These keys can be used for calling other extensions on the system (Direct Station Select or DSS keys), or can be used for options from speed dialing numbers to controlling features such as Do Not Disturb. Many features use an indicator to show whether a feature is enabled. Button programming is done through the IP Office Manager application as part of the system configuration, although some phones allow the user to program buttons and functions where given administration rights.

Busy Lamp Field (BLF) Indicators

Feature

- Status indicators which show the status of a programmable buttons associated feature or function.

Benefit

- Indication of when a button or associated feature is active.

Description

Avaya digital and IP phones have programmable buttons which can be assigned to various features. When those buttons include some form of BLF indicator, the button can also be used to indicate when the feature is active. For example, a button associated with another user will indicate when that user is active on a call. A button associated with a group will indicate when the group has calls waiting to be answered.

The speed dial icons within the IP Office Phone Manager and SoftConsole applications also act as BLF's. When the icons are associated with internal users, the icons will change to indicate the current status of the users.

Phone Manager and SoftConsole show these conditions:

- **Busy**
- **Message**
- **Forward All**
- **Do Not Disturb**

This feature is supported across the IP Office Small Community Network.

Call History

Feature

- Storage of called and calling number details within the user's phone and/or IP Office application.

Description

Most Avaya digital and IP phones keep a record of calls made and received, including unanswered calls. The method of operation varies according to the phone type but in all cases the call records can be used for return calls.

The IP Office Phone Manager application maintains a call history record of the users last 100 calls. The application must be running to record call history. Phone Manager Lite can display call history for all calls and missed calls only. Phone Manager Pro can display call histories for all calls, missed calls, inbound calls and outbound calls. Entries in the call history can be used for return calls, sorted and added to the Phone Managers local directory or speed dials.

© 2008 AVAYA All rights reserved.

Language

Avaya digital and IP phone menus and displays are available in many languages and usually the system default setting will be applicable to all phones, however it is possible to have language set on an extension by extension basis, this will also change the language of menus for IP Office Voice Mail.

Directory

The IP Office Directory is a list of up to 1000 numbers and associated names stored centrally in the system. A Directory Entry can be used to label an incoming call on a caller display telephone or on a PC application. The Directory also gives a system wide list of frequently used numbers for speed dialling via Phone Manager or a feature phone with a suitable display.

For example "Head Office" can be displayed when a known Caller ID is received. A user can also select "Head Office" in the Directory List in Phone Manager or on the display phone Directory to speed dial this number. IP Office's Directory is LDAP (Lightweight Directory Access Protocol) compliant which allows it to be synchronized with the information on any LDAP server. A maximum of 500 records can be retrieved by this method.

Self-Administration

The IP Office administrator may give select users the ability to change some of the phone settings themselves. For example, button programming. The range of changes that the user can make depends on the phone type in use.

On Hook Dialing

Avaya digital and IP phones allow the user to make calls by just dialing the number on the keypad, without having to lift the handset or pressing a speaker button. Usually the call progress can be monitored using the speaker in the phone, on phones that support hands free the whole conversation can be had without having to lift the handset.

© 2008 AVAYA All rights reserved.

Inbound Call Handling

IP Office offers several features to provide versatile inbound call processing, including PC based applications, and a standards-based TAPI interface for 3rd party applications.

Incoming Call Routing

Incoming calls can to be presented to an Operator who then decides where to pass the call, but IP Office supports intelligent call routing capable of making routing decisions based on a number of criteria.

The system currently supports routing based on;

- Call presentation digits from the exchange such as DDI/DID or ISDN MSN.
- Calling telephone number or Caller ID (This could even be part of the number received such as an area code).
- ISDN sub-address.
- ISDN/PRI service type i.e. Voice Call, Data Call, etc.

It is even possible to look for multiple criteria so, for instance, a DDI/DID call to a sales group could be handled differently depending on which part of the country the call is originating from.

Each incoming Call Route also supports a secondary destination 'Night Service' that can provide alternative routing for an incoming call based on 'time of day' and 'day of week' criteria, as well as calendar-based routing for specific dates.

Calls that cannot be routed to the configured destination are re-routed to a user defined 'Fall Back' destination. This can be particularly useful where calls are normally answered by an auto-attendant and a network fault occurs.

Where multiple call routes are set up to the same destination, a Priority level can be associated with the call. This priority level is used to determine a call's queue position in place of simple arrival time, but note that calls already ringing a free extension are not considered queuing and are not affected by a high priority call joining a queue (unless the option" Assign Call On Agent Answer" is selected for that Hunt Group). In Release 4.2 a new Priority Promotion Timer can be configured to increase the priority of calls which have been in the queue for more than a defined time.

An optional tag can be added to calls on the Incoming Call Route, which can be displayed on the alerting telephone.

© 2008 AVAYA All rights reserved.

Hunt Groups

A Hunt Group is a collection of users, typically users handling similar types of calls, e.g. a sales department. An incoming caller wishing to speak to Sales can ring one number but the call can be answered by any number of extensions that are members of the Hunt Group.

Four modes of call presentation are supported on IP Office;

- **Sequential**
  One extension at a time sequentially always starting at the top of the list.
- **Collective**
  All extensions in the Hunt Group simultaneously.
- **Rotary**
  Start with the extension in the list immediately following the extension that answered the last Hunt Group call.
- **Longest Waiting**
  Start with the extension that has been free for the longest time.

In Release 4.2 a new option has been added "Assign Call On Agent Alert" which means that CTI applications will always correctly report the details for the call that is alerting. The previous (default) behavior is now termed "Assign Call On Agent Answer" and ensures that the call at the head of the queue is always answered first.

If all extensions in the Hunt Group are busy or not answered, another Hunt Group, called an Overflow Group, can be used to take the calls. An overflow time can be set to stipulate how long a call will queue before being passed to the Overflow Group (in Release 4.2 this can be configured either for individual calls or for all calls in the group). The system can change the status of users who do not answer a hunt group call presented to them. The user can be put into busy wrap-up, busy not available or logged off. The change of status can be set per user and the use of this option can be set per hunt group.

Outside normal operation a hunt group can be put into two special modes; Night Service and Out of service.

In Night Service calls are presented to a Night Service Group. This can be controlled automatically by setting a time profile which defines the hours of operation of the main group or manually using a handset feature code.

Night service fallback using a time profile is no longer applied to a hunt group already set to Out of Service.

The Out of Service mode is controlled manually from a handset. While in this mode calls are presented to the Out of Service group

Voicemail can also be used in conjunction with Hunt Groups to take all group related messages, play an announcement when the Hunt Group is in Night Service or Out of Service mode and give announcements while a call is held in a queue. For internal voicemail use a broadcast option is provided. This feature will alter the voicemail box operation so that the message notification will only be turned off for each hunt group member when they retrieve their own copy of the message.

Small Community Networking (SCN) Distributed Hunt Groups

Small Community Networking (SCN) Distributed Hunt Groups

Hunt groups in a Small Community Network can include members located on other systems within the network. This feature requires entry of an Advanced Small Community Networking license in each system in the network.

Note: Distributed Hunt Groups are not supported for use with CCC.

© 2008 AVAYA All rights reserved.

Night Service

When a Hunt Group is in Night Service mode the Hunt Group is temporarily disabled. Callers to this Hunt Group will:

- Pass to a Night Service Fallback group used to provide cover, e.g. pass calls to a manned extension or an external number
- Be played the Out of Hours greeting if Voicemail is operational
- Receive the busy tone

A Hunt Group can be switched in or out of Night Service mode by a user dialing the appropriate short code – by any extension or by specific users.

Time Profiles

Time Profiles can be used to define when a Service, Hunt Group, Least Cost Route, Conference Bridge or a user's dial-in facility are operational. For example, a time profile can be used to route Hunt Group calls to a manned extension or voicemail outside of office hours, or be used to apply different Least Cost Routes at varying times of day to take advantage of cheaper call rates. Multiple Time Entries can be created so that a Time Profile can be used to define specific hours in the day e.g. 09:00-12:00 and 13:00-17:00. Outside of a Time Profile, voice calls would be re-routed according to the configuration but any currently connected calls at the time the Time Profile changes would not get cut off as the change only affects the routing. Data calls will get cut off as the time profile goes out of service but a new data call will start immediately if specified. From Release 4.1, Time Profiles can also be based on specific calendar dates to make allowance for public holidays or other events.

Queuing

Queuing allows calls to a Hunt Group to be held in a queue when all extensions in the group extension List are busy. When an extension becomes free the queued call is then presented. The definition of queued calls now includes ringing calls and calls waiting to be presented for ringing. The queue limit can be set to control the maximum number of calls to wait against a hunt group.

While queuing, if Voicemail is operational, the caller will be played the announcements for this Hunt Group.

Announcements

From IP Office 4.0 onwards, Hunt group announcements are separated from hunt group queuing and can be used even when queuing is off. Hunt group announcements are now supported by Embedded Voicemail in addition to VoiceMail Pro and VoiceMail Lite.

Further, times for the first announcement, second announcement, and between repeated announcements are configurable.

© 2008 AVAYA All rights reserved.

Contact Center Features

Contact Centers have specific needs for reporting on how calls are handled and these are covered in a separate section of the Product Description. Basic handling of telephony requirements for a Call Center is a standard part of IP Office from Automatic Call Distribution (ACD) and Call Queuing to agents logging on and selecting the groups that they service.

Login

A contact center agent function, login is required before the agent is able to make or receive calls from their phone. A login idle period can be specified which will dictate how long an extension can be idle before the user is automatically logged off, ensuring that an extension is not left logged in and calls go unanswered.

Monitor Calls

A user can monitor other peoples' calls by listening in. This feature is not available by default; it must be specifically enabled in the system configuration. An option exists to have a beep tone indicate when monitoring is in use. The user is only able to listen; they cannot speak into the conversation being monitored.

Acquire Call

Feature

- Takeover a call currently connected at another extension. This feature is also known as "Call Steal".

Benefit

- Assist a colleague who indicates they want you to take the call.

Description

The Acquire Call function can be setup as a special short code or programmed against a button on an Avaya digital or IP phone with programmable buttons. Use of the feature is subject to IP Office intrusion control settings, the user acquiring the call must be set to be able to intrude and the user whose call is being acquired must be set to can be intruded. Acquire call works in two ways, invoked with or without a number:

Without a value in the number field

- This allows a user to reclaim a call that was ringing on their phone but has now gone elsewhere, for example to Voicemail or Forward No Answer destination. The Intrude settings are not checked and the call can be reclaimed even if it has been answered.
- If the last call to ring this User is no longer ringing or connected on the system, the feature will fail.

With a number, where the number is the telephone number of a user who currently has the call to be acquired.

- If the user has a call ringing or waiting Acquire Call will act like the Call PickUp Extension short code and the user executing Acquire Call will be connected to the oldest ringing/waiting call.
- If the User has a connected call with no call waiting and the Intrude settings of the two Users allow it, the call will be connected to the user executing the Acquire Call and the other user will be disconnected.
- If the User does not have a call the feature will fail.

© 2008 AVAYA All rights reserved.

Queue Threshold Alert

When the number of calls queued against a Hunt Group exceed a threshold, the system can be configured to alert at a selected analog extension port. Typically the User to Alert will be a loud ringer or other alerting device. The alert does not present a real call and if answered the phone presents dial tone.

© 2008 AVAYA All rights reserved.

Miscellaneous Features

Conference Calls

Calls can be placed on hold and a conference created using either the phone or desktop applications. Additional conference members may be added up to a maximum number of 64 members.

The IP Office - Small Office Edition supports 24 conference parties with a maximum of 6 parties in any single conference.

The IP406 and IP500 can support multiple conference calls totaling up to 64 parties. For example one conference of 64 calls or 21 conferences of 3 calls each.

The IP412 has two 64-party conference bridges giving any combination from 2 x 64-party conferences to 42 x 3-party capacity.

Only two calls connecting through analog trunks are permitted in any single conference.

For more information on managing conference calls, refer to Chapter 12 where IP Office Conferencing Center is described

Dial On Pickup

Also known as "Hotline". Automatically dials a specified extension when the phone is taken off hook. This facility is commonly used in unmanned reception areas or for door entry systems to allow visitors to easily gain assistance.

Off Hook Operation

Off-Hook Station is designed for users who want their analog phone to operate like digital or IP feature phone, to isolate the user's phone idle state from the Hook state. This is a useful feature when using Phone Manager or SoftConsole to control the phone state when using a headset on an analog telephone and with call control and dialing from Phone Manager or SoftConsole.

External Control Port

The IP Office system unit has two electronic switches, similar to relays, which can be normally open, normally closed, pulsed open or pulsed closed and activated by dialing a short code or through Phone Manager, SoftConsole or VoiceMail Pro action.

These switches can be used for several purposes, for example as a means to control an electronic door release. The External Control Port switches are used to trigger/control purpose built door release equipment which is supplied by a third party. All that needs to be done is to wire the trigger/control output of the third party device to the appropriate External Control port pins.

E911

This is a specific service for North America. When an emergency call is connected, IP Office provides calling party information to an external line interface unit. The external unit carries out a number to text translation and forwards this to the emergency services bureau so that the originating location of the call is clearly identified.

© 2008 AVAYA All rights reserved.

System Short Codes

Short Codes are used as commands the IP Office to make changes for the user, group or system, so need to set up with consideration to security. The command may need additional information included with it, such as for forward, the phone number forwarded to. Short codes are a flexible and quick way of setting up certain features. IP Office has short codes provided by default on the system, or more advanced codes that need programming by the system administrator.

The full set of short code commands are listed below; please see product configuration documents for more detail on how to set them up.

AOC Previous Call
AOC Reset Total
AOC Total
Auto Attendant
Break Out
Busy
Busy On Held
Call Intrude
Call List
Call Listen
Call Pickup Any
Call Pickup Extn
Call Pickup Line
Call Pickup Group
Call Pickup Members
Call Pickup User
Call Queue
Call Record
Call Steal
Call Waiting On
Call Waiting Off
Call Waiting Suspend
Cancel All Forwarding
Cancel Ring Back When Free
Channel Monitor
Clear Call
Clear CW
Clear Hunt Group Night Service
Clear Hunt Group Out Of Service
Clear Quota
Conference Add
Conference Meet Me
CW
Dial
Dial 3K1
Dial 56K
Dial 64K
Dial CW
Dial Direct
Dial Direct Hot Line
Dial Emergency
Dial Extn
Dial Inclusion
Dial Paging
DialPhysicalExtensionByNumber
DialPhysicalNumberByID
Dial Speech
Dial V110
Dial V120
Dial Video
Disable ARS Form
Disable Internal Forwards
Disable Internal Forward Unconditional
Disable Internal Forward Busy or No Answer
Display Msg
Do Not Disturb Exception Add
Do Not Disturb Exception Delete
Do Not Disturb On
Do Not Disturb Off
Enable ARS Form
Enable Internal Forwards
Enable Internal Forward Unconditional
Enable Internal Forward Busy or No Answer
Extn Login
Extn Logout
Flash Hook
Follow Me Here
Follow Me Here Cancel
Follow Me To
Forward Hunt Group Calls On
Forward Hunt Group Calls Off
Forward Number
Forward On Busy Number
Forward On Busy On
Forward On Busy Off
Forward On No Answer On
Forward On No Answer Off
Forward Unconditional On
Forward Unconditional Off
Group Listen Off
Group Listen On
Headset Toggle
Hold Call
Hold CW
Hold Music
Hunt Group Disable
Hunt Group Enable
Last Number Redial
MCID Activate
Mobile Twinned Call Pickup
Off Hook Station
Park Call
Private Call
Private Call Off
Private Call On
Priority Call
Record Message
Relay On
Relay Off
Relay Pulse
Resume Call
Retrieve Call
Ring Back When Free
Secondary Dial Tone
Set Absent Text
Set Account Code
Set Authorization Code
Set Hunt Group Night Service
Set Hunt Group Out Of Service
Set Inside Call Seq
Set No Answer Time
Set Mobile Twinning Number
Set Mobile Twinning On
Set Mobile Twinning Off
Set Outside Call Seq
Set Ringback Seq
Set Wrap Up Time
Shut Down Embedded Voicemail
Suspend Call
Suspend CW
Toggle Calls
Unpark Call
Voicemail Collect
Voicemail Node
Voicemail On
Voicemail Off
Voicemail Ringback On
Voicemail Ringback Off

© 2008 AVAYA All rights reserved.

© 2008 AVAYA All rights reserved.

Chapter 5.
IP Telephony

© 2008 AVAYA All rights reserved.

5. IP Telephony

Technological innovation is changing the way we communicate. This time it is coming in the form of changing the way telephone calls are transmitted. It brings with it several new capabilities that change the meaning of the phrase telephone call through the use of Voice over Internet Protocol (VoIP). Basically, VoIP means "voice transmitted over a packet data network." VoIP is often referred to as IP Telephony because it uses the IP protocols to make possible enhanced voice communications throughout the world, wherever IP connections have been delivered. IP Telephony unites a company's many locations—including mobile workers— into a single converged communications network. Telephony calls using VoIP go above and beyond what's been possible in the past. When it comes to placing telephone calls, VoIP provides a range of support services and features unequalled in the world of telephony, but above all deliver them at low cost.

How Does VoIP Work?

Voice over Internet Protocol means basically what the acronym states: Voice travels over an Internet Protocol. Internet Protocol refers to the type of rules that the network uses to send and receive signals. IP Telephony works by converting voice communications into data packets. Conveniently, it runs on the popular Ethernet LAN (local area network) technology, which currently supports over 96 percent of the worlds companies' LANs.

Circuit-switched or Time-Division Multiplexed Telephony

Before digital networking with the Internet took off, everyone had to use the "Plain Old Telephone Services" (POTS). These run over a network called the Public Switched Telephone Network (PSTN). The PSTN has been around since the telephone was invented in either analog or digital form using circuit switched technology where the telephone call gets exclusive bi-directional use of a wire – or circuit – while the call is in progress. Because the circuit is exclusive to each conversation, PSTN and private branch exchanges (PBXs) must be sized to cope with peak demand and have enough circuits available for all expected conversations. This is not a flexible approach and results in a lot of infrastructure investment that the telephone companies need to recoup, via the cost of access charges and calls. The Internet has changed this – where data services have driven down access charges and allowed voice to "travel for free" over a multipurpose data network.

Packet-Switched Telephony

Unlike circuit-switched connections, which always require use of dedicated bi-directional circuit for the duration of a call, VoIP technology has enabled telephony and other new and novel features and services to run over fixed and wireless networks including private local area networks. These newer network types use packet-switched protocols. Packet-switched VoIP puts voice signals into packets. Along with the voice signals, VoIP packets include both the sender's and receiver's network addresses. VoIP packets can traverse any VoIP-compatible network. Along the way, they can choose alternate, shared paths because the destination address is included in the packet. The routing of the packets is not dependent on any particular network route which means the network provides can provide a reliable service at a fraction of the cost of circuit switched providers.

What Advantage Does IP Office Have?

IP Office can provide support of PSTN, POTs, digital time division multiplexed phones AND digital IP phones all on the same system. This means you don't have to abandon the past to embrace the future, IP Office allows all the technologies to co-exist. IP Office connects to the PSTN and to IP trunks (the VoIP equivalent) so providing a "Hybrid" PBX function – where both legacy and future technologies can be used together to minimize operating costs and offer optimize business communications through both voice and data.

IP Office has digital telephones built on both TDM and IP technology that provide the same user interface offering a flexible choice of solution that can mix, for example TDM phones in the office and IP phones at a remote site of at home. With the choice of IP phones including real and virtual (software) phones, IP Office can take communications to a new level.

Buying IP Office allows you choice – you can use the pure POTs or the pure VoIP capabilities of IP Office, or use both at the same time to allow seamless technology transition of your business without the disruption of having to choose between them now.

© 2008 AVAYA All rights reserved.

IP Office Turns VoIP into IP Telephony

In order to make use of VoIP, IP Office uses signaling protocols called H.323 right now, and Session Initiation Protocol (SIP) which allow IP Office to establish end-to-end connections for the voice path through the IP network. It ensures each end of the connection is able to transmit and receive voice and provides the network addressing for end to end packet transmission. IP Office also allows for connecting between the different technologies by translating the signals they use, for example an analog phone may wish to connect to a VoIP destination. This requires both the signaling and voice transmission to be translated – IP Office does this easily as it contains technology elements called gateways and gatekeepers that enable translations to happen.

With a conventional telephone system you plug your analog or digital TDM telephone into an extension socket connected to your PBX or Key System. With IP Telephony you connect your digital IP telephone to your IP PBX via the LAN. There are two basic types of IP phones:

- A physical phone, which looks very similar to a standard telephone (IP Hard Phone)
- A software application (Phone Manager PC Softphone) which runs on the user's PC, allowing them to use either a headset/microphone to make/receive calls anywhere they have IP connection

IP telephony has the advantage of allowing extensions to be deployed both locally and remotely through the use of IP routing and IP VPN services.

When making use of IP telephony, there are a number of data centric considerations such as which data types have priority on the IP network when there is contention. This is set with IP/TCP "quality of service" and should not be ignored. In situations where LAN Bandwidth is limited, a quality of service capable LAN switch should be used to ensure voice packets are transmitted with the required priority on the network. If not, the conversation carried over IP appears as broken up (due to packet loss) or has unacceptable delays introduced in the conversation (latency andjitter). With IP hardphones there is need for Power over Ethernet (PoE) or "midspan power" to be provided to the phones as the IP phones are no longer powered by IP Office – a list of Avaya approved PoE options is available at the end of this section.

© 2008 AVAYA All rights reserved.

Gateways, Gatekeepers and H.323 - Technology Overview

IP Office uses the H.323 signaling protocol which has the following architectural components

- Telephones are H.323 service endpoint devices that can support Audio calls. Other types of H.323 devices can support video as part of H.323
- Gateways provide media translation to allow calls to be made to non-H.323 devices, for instance an analog telephone or the public network to connect with a H.323 device
- Gatekeepers control the call processing and security for H.323 devices
- Multipoint Connection Units (MCU) for conferences by adding together media streams

These elements are grouped together in what is known as an H.323 zone (a zone is analogous to a PABX). Each zone has a single Gatekeeper that can be considered as the brains of the system dealing with call distribution, call control and the management of resources. On power-up, IP telephones, Gateways and MCU make registration requests to a Gatekeeper which then authenticates (accepts or rejects) their request to become a member of the zone. Once accepted, a telephone wishing to make a call sends a call set-up message to the Gatekeeper which then determines how to route the call and will then send an alert to the called telephone, or if the call is to a non-H.323 telephone establish the call via a Gateway within the zone.

The design of IP Telephony systems has been driven by open standards. Digital IP Phones, Gateways and Gatekeepers all support the H.323 standard and it is this that allows devices from different manufacturers to work together. IP Office has an optional integral Gateway (Voice Compression Modules) and Gatekeeper functionality required to provide a fully functional IP Telephony solution.

© 2008 AVAYA All rights reserved.

- **Gatekeeper**
  The IP Office gatekeeper allows the registration of up to 16 IP extensions on the Small Office Edition, 190 IP extensions on the IP406, 360 IP extensions on the IP412 and 272 IP extensions on the IP500, less the number of analog and digital TDM telephones already configured on the system.
- **Gateway**
  The Voice Compression Module provides the H.323 gateway function that allows IP extensions to make calls to other non-IP devices. The maximum number of simultaneous calls is limited by the number of channels available on the Voice Compression Module. IP Office must be fitted with an optional Voice Compression Module to enable IP telephony.
- **Silence Suppression**
  Silence suppression is a technique used to make the best use of available bandwidth, such as the connection over which the caller is listening, not speaking. Silence suppression works by sending descriptions of the background noise, rather than the actual noise itself, during gaps in conversation thereby reducing the number and frequency of voice packets sent on the network. Background noise is very important during a telephone call. Without noise the call will feel very unnatural and give a perception of poor quality.
- **Compression**
  IP Office supports a wide range of voice compression standards including G.711, G.729a and G.723.1. The method of compression can be either automatically established on a call-by-call basis or be configured on an individual extension basis.
- **Fast Start**
  When fast start is supported by an IP extension, this facility reduces the protocol overhead allowing an audio path to be established more quickly.
- **Out of Band DTMF**
  When out of Band DTMF is configured on an IP extension, the extension will signal to the other end of the connection which digits need to be regenerated by a local DTMF generator on behalf of the sending IP extension. This is useful when navigating external voicemail systems and Auto-Attendants.
- **Direct Media Path**
  Direct Media Path allows the speech path between two IP extensions (after call setup) to be routed directly to each other. This allows the IP Office system to free up voice compression resources after establishing the end to end connection, allowing the resources to be used in the most efficient way.
- **Auto-Create Extensions**
  IP Office can automatically create an extension entry for new IP phones added onto the local area network. In cases where the local area network is not secure this facility can be disabled, but simplifies installation of IP telephone systems
- **Fax Transport**
  Fax Transport allows fax calls to be routed over VoIP trunks between IP Office systems on an IP network using a proprietary IP Office transport protocol. This is different from the T.38 protocol which is not supported.

LAN Switch Support

Avaya recommend the use of Extreme Alpine Series switches for IP telephony applications. For more information, contact Extreme Networks.

© 2008 AVAYA All rights reserved.

Power Options for IP Telephones

Avaya supports the IEEE 802.3af, standard for Power over Ethernet (PoE) on its range of IP telephones. With Power over Ethernet, both power and data are carried over one CAT 5 Ethernet cable. Deploying IP telephones utilizing Power over Ethernet eliminates the need for local power supplies, AC adapters and cables, and allows power to be provided from the wiring closet/switch room where it can be easily connected to a UPS system.

There are several power options, in addition to IEEE Power over Ethernet, available to customers to power their Avaya IP telephones.

- **Avaya Individual Power Supply for 4600 and 5600 phones**
  Avaya provides individual power supplies that can be used to power each IP phone which provides a single 48 volt output. The power supply can operate globally within a wide range of Alternating Current (AC) input voltages: 90 - 264 Volts Alternating Current (VAC), 47-63 Hz. This power supply has a green indicator (LED) that shows the unit has power to the PHONE socket on pins 7&8 of the CAT5 cable. This item is available in two different versions, with and without an internal battery for uninterrupted power to the phone.
- **Avaya individual Power Supply for 1700 series phones**
  The 1700 series phones have a special power connecter to allow the connection of a cost efficient power supply. These power supplies come in several versions for different countries and voltages.
- **Avaya Power over Ethernet adapter for the 1703 phone**
  The 1703 phone does not support Power over Ethernet natively. A small adapter is needed. The adapter will fit completely in the stand of the phone, offering a cost efficient option of either using a power supply or a PoE adapter.

IP Telephone Power Consumption

Measured in Watts using an IEEE 802.3af power supply at 48V. This table shows the power consumption of the phones currently shipping. Older versions of phones might have a different power requirement.

		Power Consumption (Watts)		
Telephone	**PoE Class**	**Conservation Mode**	**Typical**	**Worst Case**
4602SW, 5602SW	2	4.1	N/A	5.0
4610SW, 5610SW	2	4.0	N/A	6.0
4621SW, 5621SW	2	4.9	N/A	6.45
4625SW	3	7.8	N/A	9.42
1703	2	4.32	3.75	5.10
1708	2	4.66	3.76	5.69
1716	2	3.17	2.83	3.65
1716 with 1 BM32*	2	4.37	3.41	5.52
Gig Adapter	–	3.16	N/A	5.81

*Note: when using the 1716 phone with more then 1 BM32, an external power supply is needed.

© 2008 AVAYA All rights reserved.

Avaya Individual Power Supply

Avaya provides individual power supplies that can be used to power each IP phone which provides a single 48 volt output. The power supply can operate globally within a wide range of Alternating Current (AC) input voltages: 90 - 264 Volts Alternating Current (VAC), 47-63 Hz. This power supply has a green indicator (LED) that shows the unit has power to the PHONE socket on pins 7&8 of the CAT5 cable.

This item is available in two different versions, with and without an internal battery for uninterrupted power to the phone.

1151 local power supply, without battery and with battery backup

Avaya Mid-Span Power Distribution Units

These power devices are designed for IP-telephony and provide power over Ethernet (PoE) for up to 24 IP telephones or wireless LAN (WLAN) access points in one unit. The Mid Span Power units are designed to mount in a 19-inch rack with the data equipment or they can be stacked up to four units high using the optional rubber feet. The mid-span is 1U in height (1.75 inches) and has up to twenty-four RJ45 sockets on the bottom row and twenty-four data and power output RJ45 sockets on the top row. The units provide a maximum of 200 Watts or a peak of 16.8 watts per port. Data is unaffected by power delivery, if the device does not require power. The mid-span power units are also referred to as PDU (Powered Data Unit) devices. Power over the LAN will simplify the installation and support of IP telephones for our customers and are available in 3 sizes; 6, 12 or 24 ports with optional SNMP management capability.

Mid-Span power supply

© 2008 AVAYA All rights reserved.

VoIP FAQ

Network Requirements

Quality of Service (QoS) is a measure of the performance of a network that reflects the availability of network service and the quality of network transmissions. The term itself refers to a number of networking technologies and techniques and does not necessarily restrict itself to any single protocol or standard.

There are a number of measures that can be taken on the LAN and WAN to make them 'good enough' to carry voice traffic. Some of these are the implementation of standards based QoS protocols while are simply a matter of network architecture and good network management practices.

The term 'good enough' is intentional. Every customer will have different expectations and different budgets to work to. Some will be willing to upgrade their networks to use the best possible equipment and practices. To others the additional expense may be viewed as unnecessary.

Examples of standards based Quality of Service protocols include DiffServ and 802.1p/q.

What are Voice Compression Modules (VCM's) for?

VCMs are required to support the following scenarios:

- Usage of Embedded Voicemail on the Small Office Edition (used as a memory boost by compressing the voice files).
- Internal phone calls between an IP device and a non-IP device.
- Analog/digital phones to IP trunks (SIP/H.323) including managed Frame Relay and managed IP VPN (provides echo cancellation).
- IP phones to ISDN or PSTN trunks (convert IP to TDM and vice-versa).
- Call set up between IP phones (VCM resource will be released after call set up if direct media is used) to provide dial tone, busy tone etc. Direct media is a VoIP concept within the system that allows direct connection of the media steam (IP packets containing voice samples of the telephone call) between the two IP devices on the network.

VCMs are NOT required for:

- Calls between IP phones on the same system after call set-up ("Direct Media"), unless call recording is enabled.

"Direct Media" is a VoIP concept that circumvents resources (TDM bus, Gateway) within the system and improves the voice quality. If two IP devices are connected on the same system, a direct LAN connection between them will be established once the call has been set up (as long as they use the same Codecs).

It is possible for an IP device to be physically located at one site while being registered at a different site. In this case, even for VoIP across the WAN the VCM would not be used, as long as the two IP devices involved in a phone call are registered on the same system.

Data Channels

A Data Channel is only required for Remote Access (RAS), Internet Access, and Voicemail connections:

- 10 PCs accessing the Internet over a single line = 1 Data Channel. If multiple lines are used (Multi-Link PPP) then as many data channels are required (e.g. 128k i.e. 2B channels requires 2 data channels)
- 10 users dialing in from home on 10 separate lines onto the LAN = 10 Data Channels
- Voicemail is an IP application on the LAN (i.e. one data channel is required for each voicemail port used)

Note: IP end-points do NOT require data channels

© 2008 AVAYA All rights reserved.

Bandwidth Required For Each Voice Call?

The bandwidth used varies depending on the compression method chosen. IP Office supports a wide range of compression standards, including the most popular G.723.1 and G.729a. These will occupy approximately 10K and 13K of bandwidth respectively.

Use the following chart to choose the most appropriate compression algorithm for your available bandwidth.

Audio Codec	RTP Voice Data Payload	Packets per Second	LAN (bps)	% Overhead LAN	WAN (bps)	% Overhead WAN	Algorithmic Delay (milli-seconds)
G.723.1	24 Bytes	33.33	20,800	225%	9,867	54%	80
G.729a	20 Bytes	50	29,600	270%	13,200	65%	40
G.711 (64K)	160 Bytes	50	85,600	34%	69,200	8%	20

Acceptable Delay?

End-to-end delay should be 150 milliseconds or below.

How Many Simultaneous Calls Can I Get Down My Link?

The following chart illustrates the theoretical maximum number of simultaneous voice calls that can be delivered over a WAN for a given link speed. This does not take into account any bandwidth that may be required for data traffic between sites or the physical limit of VoIP calls for the specific version of IP Office in use.

The number of simultaneous voice calls can be in excess of the capabilities of the individual platform, where the calls transit the switch as data traffic. In this situation compression resources are not used but obviously must be catered for in the overall bandwidth provision.

Compression	G.723.1 (6K3)	G.729a (8K)	G.711 (64K)
Algorithmic Delay (seconds)	0.08	0.04	0.02
Number of Calls			
- 64Kbps Link	6	4	0
- 128Kbps Link	12	9	1
- 256Kbps Link	25	19	3
- 512Kbps Link	51	38	7
- 1Mbps Link	103	77	14
- 2Mbps Link	207	155	29

© 2008 AVAYA All rights reserved.

What is the Maximum Number of Simultaneous VoIP Calls

Each IP Office can be fitted with an optional Voice Compression Module (VCM) to support VoIP connections.

- The IP406 can be fitted with a single module offering up to 30 simultaneous calls.
- The IP412 is capable of supporting two modules of all types, allowing up to 60 simultaneous calls.
- The IP500 is capable of supporting two VCM 32/64 modules allowing up to 128 simultaneous calls.

Does the IP Office support Fax over IP ?

The IP Office has a proprietary method for carrying Fax traffic on a VoIP call. IP Office does not currently support the T.38 Fax standard. IP Office supports Fax speeds up to 14.4 Kbps. The bandwidth requirements for a Fax call will initially be as per the specified or negotiated compression method and then the bandwidth requirement will change to accommodate the Fax data. The Fax bandwidth will vary depending on the speed with which the Fax devices are communicating and the type of link, at 14.4 Kbps the bandwidth requirement will be approximately 27 Kbps on the LAN or 19 Kbps on a Point to Point WAN link with header compression enabled.

Network Assessment

With IP Office, optimum network configurations can support VoIP with a perceived voice quality equivalent to that of the Public Switched Telephone Network (PSTN). However, not every network is able to take advantage of VoIP transmissions. It is important to distinguish between basic compliance with the minimal VoIP standards and validated support for QoS which is needed to run VoIP applications over a data network.

With the exception of standalone configurations where IP phones connect directly connected to the ports on IP Office, Avaya now requires that all customers formally audit their networks for IP telephony readiness before attempting to install any VoIP application.

A network assessment should normally include:

- Physical inventory of all equipment inclusive of the current version of code, and configurations as needed.
- An accurate and complete network topology for all networked sites, inclusive of IP addressing and physical/logical connections.
- An evaluation of the network's topology to check that the design is both sound and reasonable.
- Measurement of packet loss, jitter and delay over the course of multiple days and measured on a per minute basis. A graphical representation of the data is the preferred output method.
- Examination of QoS/Class of Service (CoS) parameters in place in the network.
- Summary of findings and possible actions to correct problems.

The assessment should leave you confident that the implemented network will have the capacity for the foreseen data and voice traffic, and can support H.323, DHCP, TFTP, and jitter buffers in H.323 applications.

With this in mind, if you require support during or after an IP Office VoIP installation, a copy of your network assessment documentation will be requested by your support channel.

For more details about available tools, resources and services to enable you to audit your network for VoIP readiness, please contact your local Avaya representative.

© 2008 AVAYA All rights reserved.

IP Packet Flow Control

While a high-performance switch forwards data packets at full wire speed to and from its ports simultaneously, there may be times when a switch port may not be able to accept packets at the rate it is receiving them.

For example, the switch port may be receiving packets from multiple ports at the same time, or the switch port may be receiving packets from a port operating at a faster speed. For instance, the sending port might be operating at 100 Mbps, while the receiving port operates at 10 Mbps; or the sending port might operate at 1000 Mbps, while the receiving port operates at 100 or 10 Mbps. If data packets arrive for a port that is saturated with other packets, the packets may overflow the port's buffer, resulting in dropped packets and lost data.

Flow control is a congestion-control mechanism that prevents data loss at congested ports. Flow control prevents packet loss by controlling the flow of data from the transmitting device to ensure that the receiving device can handle all of the incoming data.

IEEE 802.3 flow control is used on Avaya IP telephones operating in full-duplex mode. If the receiving device becomes congested, it sends a pause frame to the transmitting device. The pause frame instructs the transmitting device to stop sending packets for a specific period of time. The transmitting device waits the requested time before sending more data.

VoIP Standards Supported

IP Office supports the following protocols and standards:

- H.323 V2 (1998), Packet-based multimedia communications systems.
- Q.931, ISDN user-network interface layer 3 specification for basic call control.
- H.225.0 (1998), Call signaling protocols and media stream packetization for packet-based multimedia communication systems.
- H.245 (1998), Control protocol for multimedia communication.
- Session Initiation Protocol.
- Audio CODECs:
  - G.711 A-law/U-law.
  - G.723.1 MP-MLQ.
  - G.729 Annex A – CS-ACELP.
- Silence Suppression.
- Fax Relay (IP Office to IP Office Fax Transport over IP).
- Local End Echo Cancellation 25ms.
- Out of band DTMF.
- Jitter buffer, 5 frames of jitter buffer.
- Internet Standards/Specification (in addition to TCP/UDP/IP).
  - RFC 1889 – RTP/RTCP, Real Time and Real Time Control Protocol.
  - RFC 2507, 2508, 2509 – Header Compression.
  - RFC 2474 – DiffServ, Type of Service field configurable.
  - RFC 1990 - PPP Fragmentation.
  - RFC 1490 - Encapsulation for Frame Relay.
  - RFC 2686 - Multiclass Extensions to Multilink PPP.
  - RFC 3261 - Session Initiation Protocol (SIP).

© 2008 AVAYA All rights reserved.

© 2008 AVAYA All rights reserved.

Chapter 6. Public and Private Voice Networks

© 2008 AVAYA All rights reserved.

6. Public and Private Voice Networks

With Avaya IP Office you can be networked via T1, PRI and BRI ISDN, including VoIP on the company WAN. Networking maximizes the current potential of your branch maximizes the current potential of your branch office and remote workers—while building the office and remote workers—while building the best possible foundation for your future growth. IP Office provides each location with a scalable (up to 360 users) telephony solution that supports voice networking, and offers:

- A uniform dialing plan, making it easy to call co-workers anywhere on the network and improve customer service
- Consistent user experience by sharing the same phones and messaging interface as in headquarters
- A user-defined central directory that is automatically synchronized
- Least cost routing and bandwidth on demand
- Centralized voicemail and/or the ability to network voicemail systems together

The benefits of networking:

- Operate a network of branch offices with a consistent set of communications and services across all locations; gain the efficiencies of universal functions and end-user familiarity.
- Leverage any existing investment in Avaya systems at other sites
- Centralize services (e.g. operator, voicemail) as well as management and administration to reduce costs
- Speed deployment of remote offices—respond more quickly to market demands.
- Improve inter-site communication to simplify information exchange and enhance customer service.

© 2008 AVAYA All rights reserved.

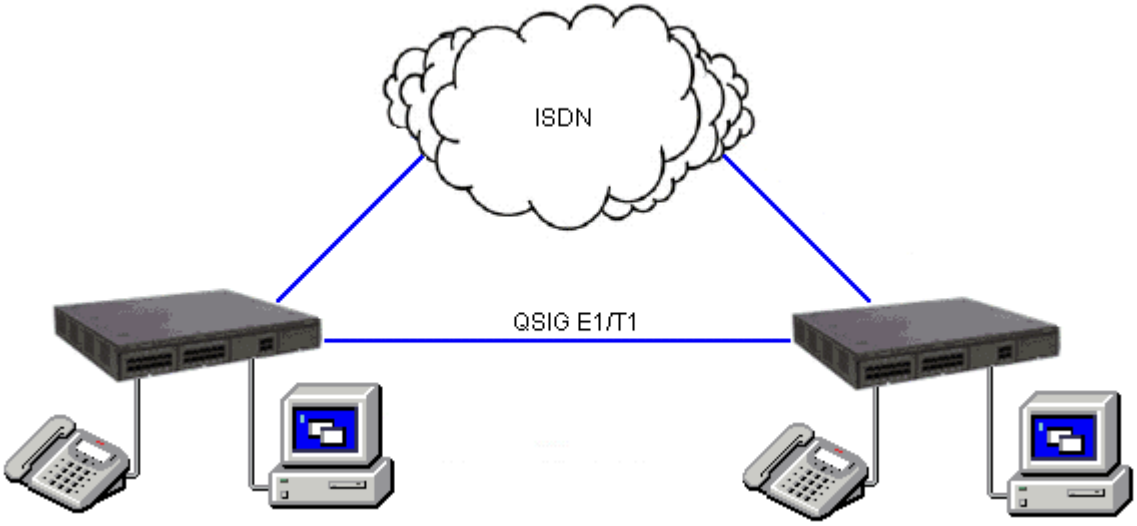
Private Circuit Switched Voice Networking

Private voice networks are built using structured leased line circuits (E1 or T1) or by establishing permanently connected 'B' channels over ISDN circuits between IP Office systems. Each channel within the E1 or T1 interface can provide a single voice or 64K/56K data call. Where leased line circuits are used within a private networking scenario, these E1 or T1 interfaces are typically configured to use QSIG signaling between sites.

QSIG provides a level of voice feature transparency between PBXs and is the favored signaling standard within multiple vendor and international voice networks. The IP Office E1 or T1 module terminates a QSIG connection with a 120 ohm RJ45 interface.

IP Office supports the following QSIG services across this network:

- **Simple Telephony Call/Basic Call:** ETS300 171/172.
- **Circuit Switched Data Call/Basic Call:** ETS300 171/172.
- **Called/Calling Line ID Presentation:** ETS300 173.
- **Called/Calling Name Presentation:** (SS-CNIP, SS-CONP, SS-CNIR) ETS300 237/238.
- **Message Waiting:** (SS-MWI) EN301 260/255.
- **Transfer:** (SS-CT) ETS 300 260/261.



Circuit Switched Voice Networking

© 2008 AVAYA All rights reserved.

The IP Office platform supports a range of trunks and signaling modes for connection to the public switched telephone network (Central Office). Some of these lines are only available in certain territories; please check with your distributor for local availability. Primary rate trunks are available with either a single (24/30 channels) or dual trunk (48/60 channels).

ISDN Primary Rate (ETSI CTR4)

Provided by the IP400 PRI E1 and IP500 Universal PRI cards.

ISDN Primary Rate provides 30 x 64K PCM speech channels over an E1 circuit and one signaling channel. Signaling Conforms to the ETSI Q.931 standard with Cyclic Redundancy error Checking (CRC).

The following supplementary services are supported:

- Calling Line Identification Presentation (CLIP) provides the telephone number of the incoming call to the IP Office.
- Calling Line Identification Restriction (CLIR) prevents the telephone number of the IP Office being presented on an outbound call.
- Connected Line Identification Restriction (COLR) Inhibits the COLP service.
- Direct Dialing In (DDI) where the exchange provides the last x digits of the dialed number on an incoming call. This allows IP Office to route the call to different users or services.
- Sub-addressing Allows the transmission/reception of up to 20 digits, additional to any DDI/DID or CLIP information, for call routing and identification purposes.

ISDN Basic Rate (ETSI CTR3)

Provided by the IP400 Quad BRI and IP500 BRI cards.

ISDN Basic rate provides 2 x 64K PCM speech channels and one signaling channel using Q.931 signaling and CRC error checking. Both point to point and point to multipoint operation is supported. Multipoint lines allow multiple devices to share the same line; however point-to-point is the preferred mode.

Basic rate supports all the services that are supported on the primary rate version with the addition of

- Multiple Subscriber Number. This service is usually mutually exclusive with the DDI/DID service and provides up to 10 numbers for routing purposes, very similar to DDI/DID.

© 2008 AVAYA All rights reserved.

Additional ISDN features

The following ISDN features are supported by IP Office 4.0+ on both PRI and BRI trunks. Note that availability of these features is dependant on their also being supported and available from the ISDN service provider, for which there may be charges.

- **Malicious Call Identification – MCID**
  (24xx, 46xx, 54xx, 56xx, T3, T3 IP, DECT phones; Phone Manager)
  Short codes and button programming features are available so that users can trigger this activity at the ISDN exchange when required. This feature is NOT available on standard ISDN DSS1 phones.

- **Advice of Charge – AOC**
  (T3 digital and IP phones only; Phone Manager)
  Advice of charge during a call (AOC-D) and at the end of a call (AOC-E) is supported for outgoing ISDN calls other than QSIG. The call cost is displayable on T3 phones as well as Phone Manager and included in the IP Office Delta Server SMDR output for call accounting purposes. The IP Office allows configuration of call cost currency and a call cost mark-up for each user.

- **Call Completion to Busy Subscriber – CCBS**
  (2400, 4600, 5400, 5600, T3, T3 IP, DECT phones; Phone Manager)
  CCBS can be used where provided by the ISDN service provider. It allows a callback to be set on external ISDN calls that return busy. It can also be used by incoming ISDN calls to a busy user. This feature is NOT available on standard ISDN DSS1 phones.

- **Partial Rerouting – PR**
  (2400, 4600, 5400, 5600, T3, T3 IP, DECT phones; Phone Manager)
  When forwarding a call on an ISDN channel to an external number using another ISDN channel, partial rerouting informs the ISDN exchange to perform the forward, thus freeing the channels to the IP Office. This feature is NOT available on standard ISDN DSS1 phones and it is NOT supported on QSIG.

- **Explicit Call Transfer – ECT**
  (The normal usage of this feature is by a third party application)
  ECT is supported on the S0 interface. A Call to an S0 Endpoint can be transferred to any other device such as an analog, digital or IP endpoint or to any trunk. The normal usage of this feature is by a third party application connected via one or more S0 interfaces to IP Office. One example is the VoiceDirector, an automatic call assistant.

© 2008 AVAYA All rights reserved.

North American T1

Provided by the IP400 PRI T1 and IP500 Universal PRI cards.

T1 Primary Rate provides up to 24 64K channels over a 1.54M circuit. Each channel of the T1 trunk can be independently configured (channelized) to support the following signaling emulations with handshake types of immediate, delay or wink.

- Loop-Start
- Ground-Start
- E&M Tie Line
- E&M DID
- E&M Switched 56K
- DID - Channels configured for DID/DDI support incoming calls only. The carrier or Central Office will provide the last x digits that were dialed to be used for call routing.
- Wink-Start

IP Office T1 trunks support both DNIS and ANI services, where available from the central office.

- Dialed Number Identification String (DNIS) Provides a string of digits to the IP Office depending on the number dialed by the incoming caller. This string can then be used to route callers to individual extensions, groups or services.
- Automatic Number Identification (ANI) Provides IP Office with a number identifying who the caller is. This may then be used for routing or computer telephony applications.

T1 trunk cards incorporate an integral CSU/DSU, eliminating the need for an external unit. The CSU function allows the trunk to be put in loop-back mode for testing purposes. This can be set manually, using the monitor application, or automatically from a Central Office sending a Line Loop Back (LLB) pattern. The DSU function allows the T1 trunk to be shared between data and voice services.

North American Primary Rate Interface

Provided by the IP400 PRI T1 and IP500 Universal PRI cards.

IP Office supports Primary Rate ISDN trunks on 5ESS or DMS100 central office switches provided by AT&T, Sprint, WorldCom and other Local Telcos. Channels can be pre-configured for the supported services or negotiated on a call-by-call basis.

Special Services can be configured to route calls to local operators or pre-subscribed carriers for both national and international calls (SSS). Alternate carriers can also be selected through the configuration of IP Offices Transit Network Selection (TNS) tables.

IP Office also supports the Calling Name and Number service over Primary Rate Trunks (NI2).

© 2008 AVAYA All rights reserved.

Analog Trunks

- **Loop Start**
  Loop start trunks are available on the IP Office Analog Trunk cards installed within the IP Office control unit, or on the Analog Trunk 16-port expansion modules (ATM16). The first two trunks on the ATM16 are automatically switched to power fail sockets in the event of power being interrupted. They conform to the TIA/EIA-646-B standard. The loop start trunks also support incoming caller line identification (ICLID) conforming to GR-188-CORE and GR-31-CORE standards. IP Office can use this information to route calls or provide it to computer applications to display additional information about the caller.
- **Ground Start**
  Ground Start trunks are only available on the ATM16, configured through IP Office Manager. The first two trunks on the module are automatically switched to power fail socket in the event of power being interrupted. They conform to ANSI T1.401 and TIA/EIA-646-B standards. Not available in all territories.

E1R2 Channel Associated Signaling

Provided by the IP400 PRI E1R2 and IP500 Universal PRI cards.

The IP400 PRI E1R2 cards are available in two versions supporting either RJ45 or coax network connections. Each card provides channels that can be configured for MFC, Pulse or DTMF dialing dependent on the requirements of the network.

© 2008 AVAYA All rights reserved.

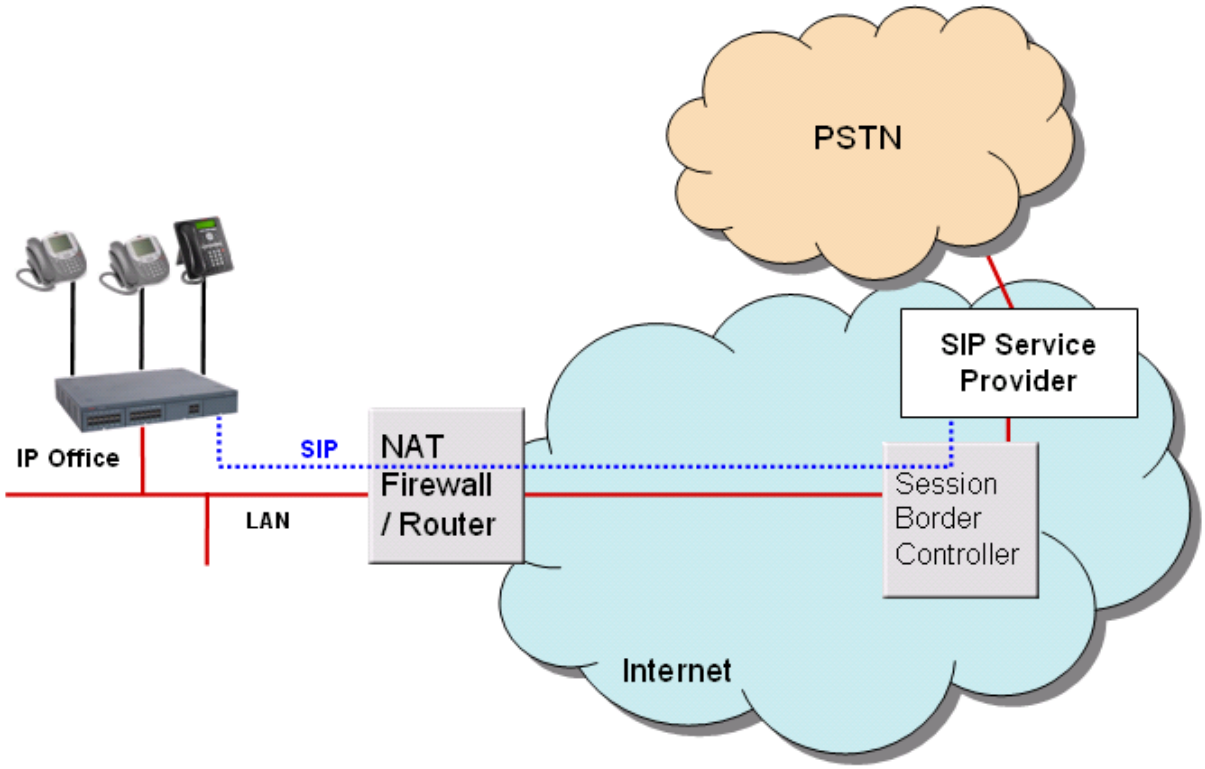
Session Initiation Protocol (SIP)

IP Office 4.0+ introduces SIP trunking. SIP trunks allow IP Office users to take advantage of new telephony services being offered by 'Internet Telephony Service Providers (ITSPs)'. In many cases, these telephony services can offer substantial savings in comparison to traditional exchange lines. The IP Office solution allows all users, regardless of their phone type, to make and receive SIP calls. SIP trunks are handled like any other line on IP Office, affording all the call routing and toll control needed to manage inbound and outbound calls.

SIP trunks on IP Office require the provisioning of voice compression channels through the installation of VCM modules within the control unit. RTP Relay is also supported to allow the IP stream through SIP after call setup. A license for the maximum required number of simultaneous SIP calls is also needed.

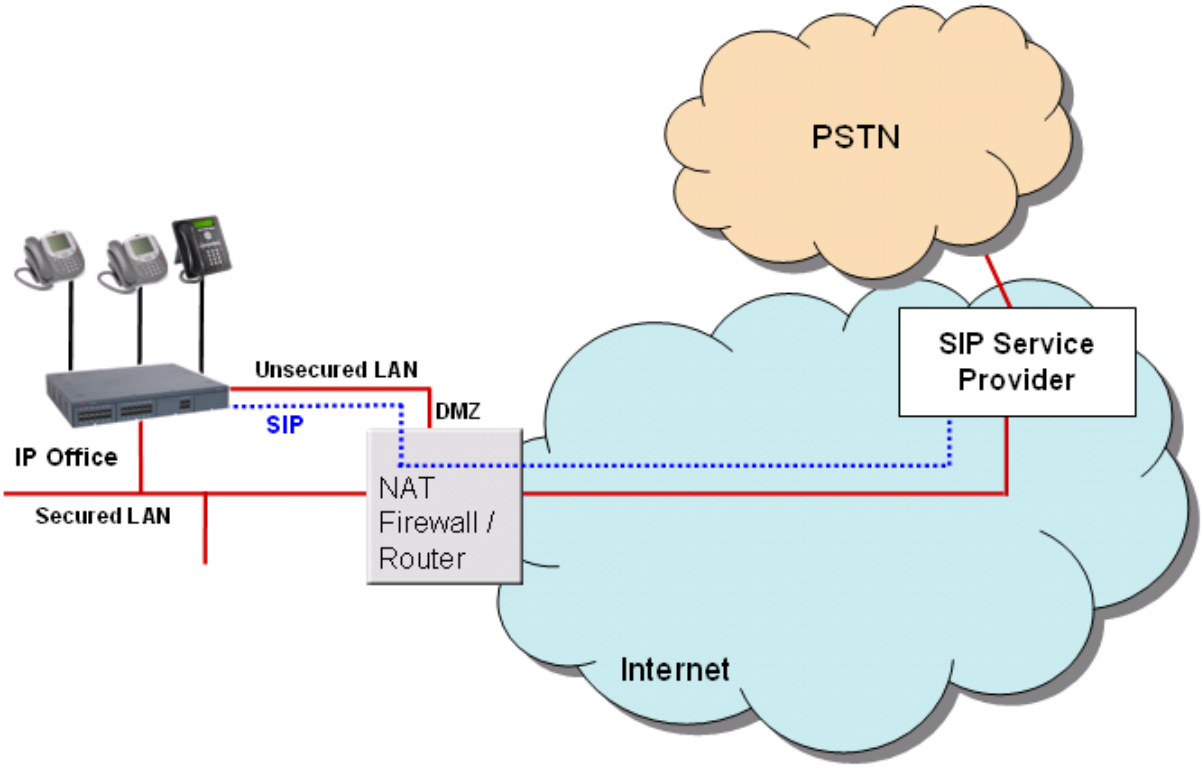
There are several possible network topologies for SIP trunk systems, as shown in the following diagrams.

Option 1: Service provider with a Session Border Controller (SBC), which solves NAT traversal issues – this is the most reliable and preferred method.

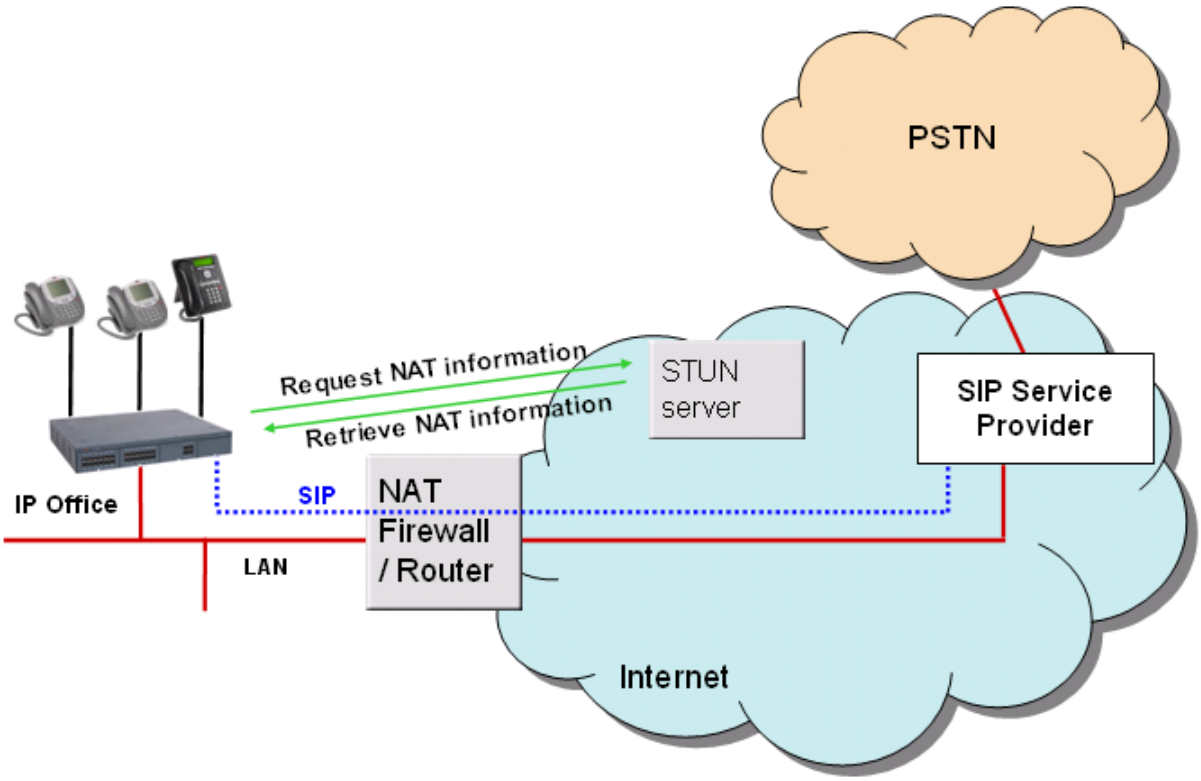


© 2008 AVAYA All rights reserved.

Option 2: Direct connection from the IP Office's second Ethernet port to the internet via a DMZ (demilitarized zone) port on the router. To make this configuration secure, the IP Office firewall is set to drop all packets except SIP.

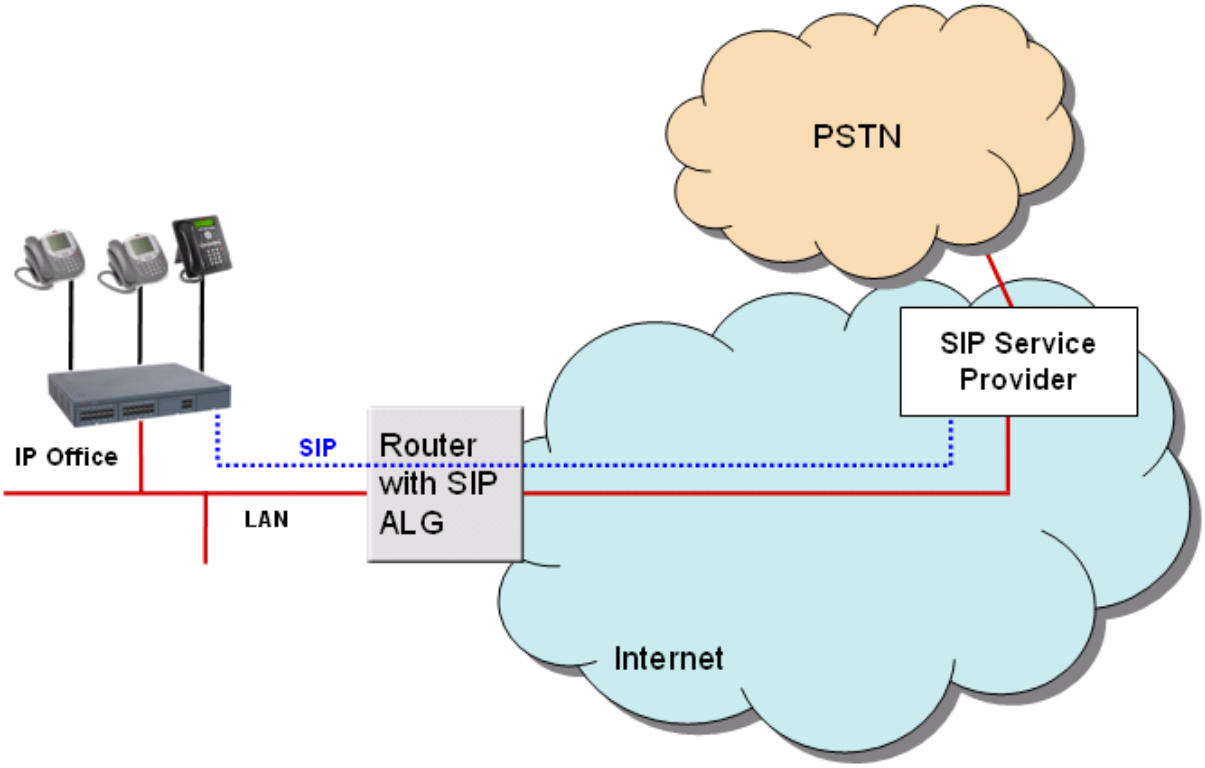


Option 3: Connection to the ITSP over NAT using 3rd party STUN (Simple Traversal of UDP through NAT) servers in the network to discover the NAT mechanism being used.



© 2008 AVAYA All rights reserved.

Option 4: Connection to the ITSP through a router equipped with an Application Level Gateway (ALG) which transparently resolves all NAT issues.



For details on SIP ITSPs which have been tested by Avaya, please see the Technical Bulletin for the IP Office 4.0 release and/or IP Office Knowledge Base at http://www.avaya.com/ipoffice/knowledgebase.

© 2008 AVAYA All rights reserved.

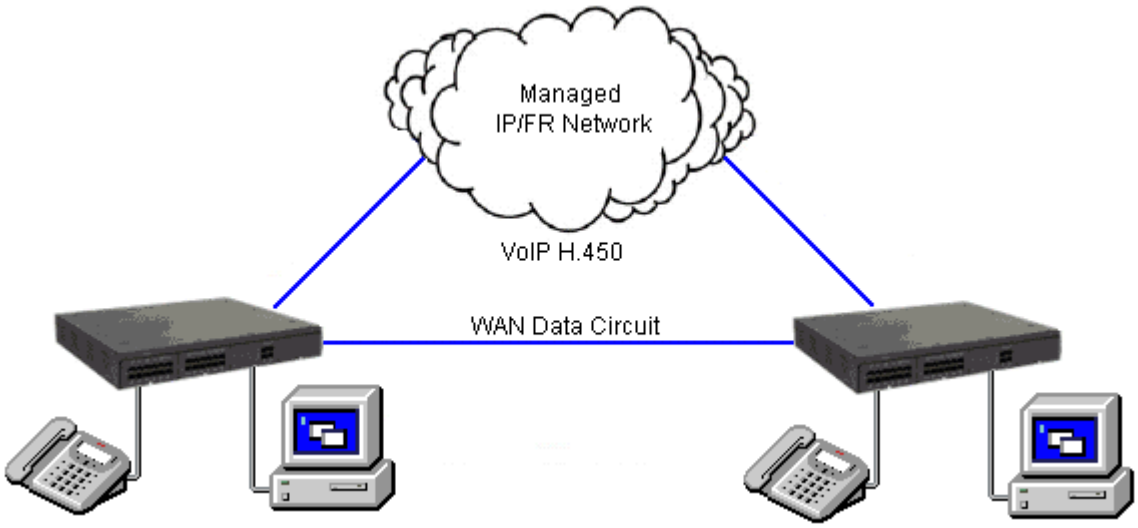
Packet Based Voice Networking

This section describes the options available for businesses that are able to utilize data networks to support voice solutions such as Voice over IP (VoIP). IP Office offers networked voice and data communications, providing:

- Built-in IP router
- One link for voice and data networking
- Common access to the Internet; share files and send e-mails to other sites
- Support for RIP-2 protocol for dynamic data routing; IPSec VPN, firewall and NAT (Network Address Translation) and for security; Centralized management and proactive fault management via SNMP.

IP Packet based voice networking between IP Office sites can be achieved in a number of ways:

- VoIP over an unstructured private circuit.
- VoIP over a managed IP VPN.
- VoIP over a managed Frame Relay network.
- VoIP across the campus LAN.
- VoIP across the public network.



VoIP networking across IP network or WAN

VoIP over an Unstructured Private Circuit

Data networks can be constructed with IP Office using unstructured point-to-point data circuits (X.21, V.35) at speeds of up to 2 Mbps. These data circuits are accessed via optional Wide Area Expansion modules (one port is included on IP400 system units) and Voice Compression Modules (VCM). This approach can realize significant savings by allowing packetized VoIP calls to be interleaved with data on up to 7 leased data circuits with spare bandwidth. Depending on required solution sizing, IP Office supports from 3 to 128 concurrent VoIP calls. This option is not supported on IP500 systems.

© 2008 AVAYA All rights reserved.

VoIP over a Managed Frame Relay Network

Frame Relay is a high-speed, packet switching WAN protocol that enables the interconnection of LANs and is usually offered as a service by a public network provider. Frame Relay is a connection-oriented protocol, which means that it relies on an existing end-to-end path between devices connected across the network. It implements these connections using Permanent Virtual Circuits (PVCs).

Like a leased circuit, a PVC is a logical path that connects two devices. This path between the source and destination point is a dedicated connection, so the PVC is always available to the connected devices. However, unlike a leased circuit many PVCs can coexist on a single access circuit which allows devices to share the bandwidth of a given transmission line.

Voice over a managed Frame Relay network is similar to Voice over a managed IP network except that the access interface is usually an unstructured leased circuit via IP Office's WAN port. IP Office employs a Frame Relay Assembler Disassembler (FRAD) to allow voice and data traffic to be formatted and framed for a Frame Relay network.

VoIP over a Managed IP VPN

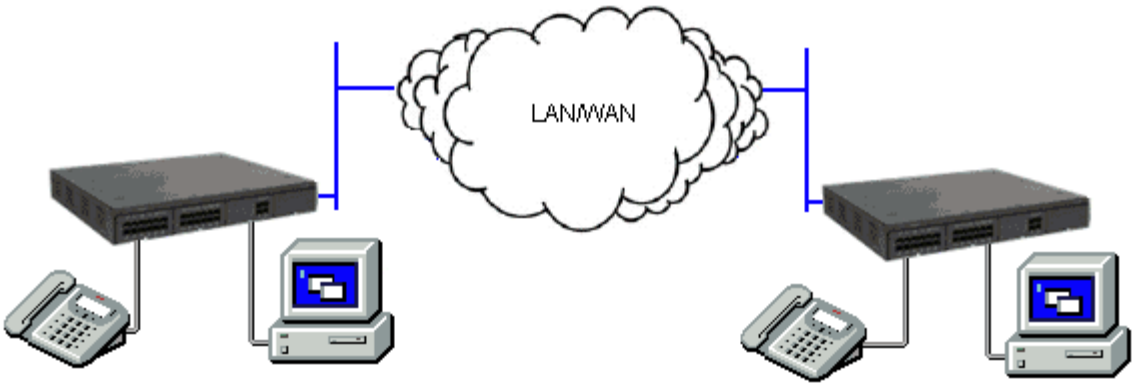
Even though IP Office can operate as a pure circuit switched system with analog and digital TDM handsets, because IP Office includes an integrated Voice over IP (VoIP) Gateway significant cost savings can be made by sending voice and data over a single managed IP VPN.

IP VPNs have advantages over Frame Relay networks; access bandwidth need not be pre-allocated between sites like Frame Relay's PVCs and they are generally lower cost and their global reach is normally greater. Access to the IP VPN is via one of IP Office's WAN ports.

A managed IP network or IP VPN is a private network of routers managed and partitioned by a single network service provider who assigns IP addresses and manages the network. Because of this, the network service provider can guarantee throughput levels, minimize latency and ensure transmission speeds to give greater quality of service supported by a contracted service level agreement. Avaya do not recommend networking IP Office systems over a unmanaged public IP networks where neither QoS nor service levels can be guaranteed by the provider.

VoIP across the LAN

In a factory or campus environment, voice calls can sent over 10/100 Mbps LAN connections on systems equipped with optional Voice Compression Modules (VCM). In order to avoid bandwidth contention issues, VoIP across the LAN will require some form of bandwidth management through Diffserve.



VoIP networking across the LAN

VoIP across the Public Network

IP Office is capable of realizing the benefits of Q.931 and H.450 supplementary service support across a public connection where an appropriate QoS connection can be established.

© 2008 AVAYA All rights reserved.

Supplementary Services within IP Networks

Supplementary services within an IP environment are provided via Q.931 and H.323. IP Office provides the same rich services as enjoyed within a traditional network environment. Our standards based approach allows interoperability within mixed vendor networks.

The basic supplementary service features supported by H.323 on IP Office to IP Office IP trunk links are listed below.

- **Basic call set up (voice).**
- **Call Hold (local).**
- **Call Transfer (local).**
- **Called/Calling Name.**
- **Called/Calling Number.**

Additional features can be added through the use of IP Office Small Community Networking (see the following section).

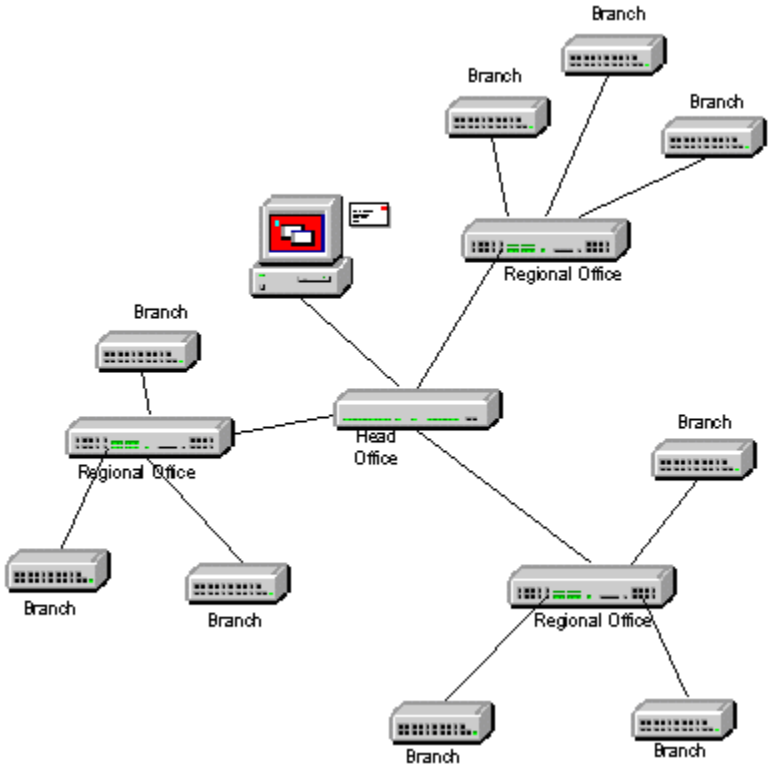
On IP trunks to non-IP Office systems the Supplementary Service will depend on those also supported by the non-IP Office system.

© 2008 AVAYA All rights reserved.

Small Community Networking

When connecting IP Offices together over IP or Packet based networks, Small Community Networking enhances feature transparency. These networks can support up to a maximum of 500 users across 16 sites. The following additional features are available.

- **Busy Lamp Field.**
- **Camp-on.**
- **Call Back When Free.**
- **Paging.**
- **Call Pick-up.**
- **Centralized Voice Mail (VoiceMail Pro)**
  Support for mailboxes, call recording, dial by name and auto attendants. Remote queuing on remote systems is also supported with the Advanced Small Community Networking license (see below).
- **Internal Directory.**
- **Absent Text Message.**
- **Anti-Tromboning.**



For Small Community Networks VCM modules are required in all systems being connected. The IP lines should be configured to connect the IP Offices in a star configuration, however the data network itself can be meshed. Also the names and numbers (groups, line, services, etc) on the separate IP Office systems should be unique to reduce potential maintenance confusion.

Each IP Office system broadcasts UDP messages on Port 50795. These broadcasts typically recur every 30 seconds but BLF updates are potentially more frequent. There are no updates if there is no activity and the overall level of traffic is very low – typically less than 1 kbps per system.

From IP Office Release 2.1(35) and higher, SCN is supported between IP Office systems with differing software levels but network features will be based on the lowest level of software within the network. This option is intended to allow the phased upgrading of sites within a SCN and it is still recommended that all systems within a network are upgraded to the same level where possible.

If larger networks are required QSIG can be used to link multiple Small Community Networks together. Functionality between the communities is governed by the QSIG feature set.

Note: On IP500 systems, Small Community Networking requires one or more additional licenses.

© 2008 AVAYA All rights reserved.

Small Community Networking - Advanced Networking Features

IP Office R4.0 allows a number of additional feature to be enabled by addition of an Advanced Small Community Networking license. Those features are:

- **Distributed Hunt Groups**
  Hunt groups can include users located on other IP Office systems within the network.
- **Remote Hot Desking**
  Users can hot desk between IP Office systems within the network. The system on which the user configured is termed their 'home' IP Office; all other systems are 'remote' IP Offices. To log on at a remote IP Office requires that IP Office to have an Advanced Small Community Networking license. A license is not necessary on the user's home IP Office.
- **Breakout Dialing**
  This feature allows the user to select an IP Office system in the network from a displayed list and then dial a subsequent number as if dialing locally on the select system. This feature is triggered either by a programmable button or short code.

Note that both Distributed Hunt Groups and Remote Hot Desking are not supported for use with CCC.

The Advanced Small Community Networking license is required in every IP Office site where remote workers are expected to hot desk to as well as on every site where members are included in distributed groups.

© 2008 AVAYA All rights reserved.

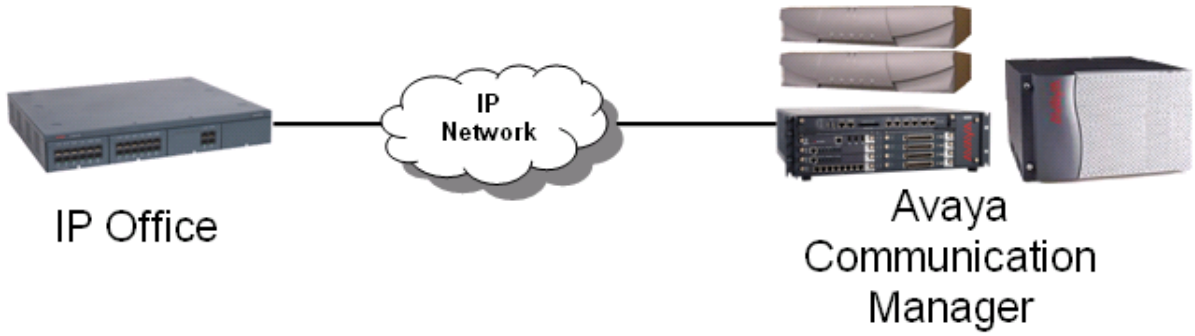
Internetworking with Other Avaya Products

IP Office will support the most appropriate way for communication with any other existing PBX's in a customer network, whether TDM or IP-based. With Avaya DEFINITY, MultiVantage, Avaya Integral Enterprise or Avaya Communication Manager (ACM), the protocols used will be QSIG or H.323 over T1, E1 or IP links

VoIP networking using H.323

IP Office (since release 1.1 in US and release 1.2 in EMEA) has been successfully tested to be interoperable over IP trunks with DEFINITY G3si (release 10) and IP600 (release 9.5). The protocol supported is H.323 using industry-standard compression codecs (types G.711A, G.711MU, G.729A and G.723.1-6K3). The features currently supported are as follows:

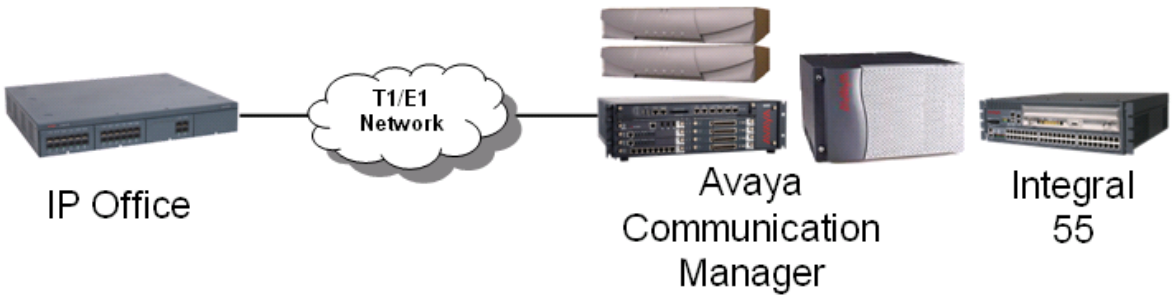
- Desk to desk dialing (basic voice call)
- Calling/Connected Party ID number
- Calling/Connected Name Presentation
- Call Hold
- Call Transfer



These features allow for simple networking needs between IP Office remote branches to a DEFINITY/ACM at the main site.

QSIG networking using T1/E1 links (TDM)

Alternatively QSIG may be favored as the chosen signaling standard within multiple vendor environments and provides the following supplementary services which are also available between IP Office and DEFINITY / MultiVantage/ I55 /ACM (equipped with the relevant RFA licenses):



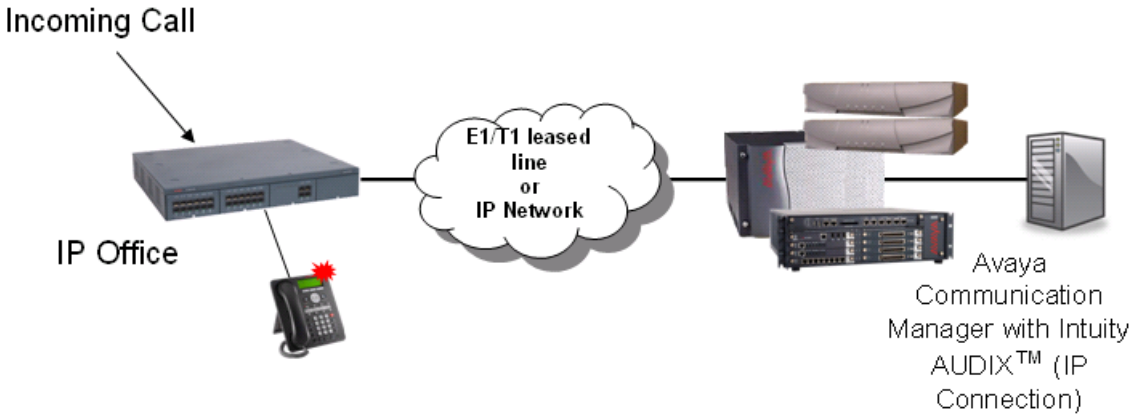
- Simple Telephony Call/Basic call (ETS 300 171/172)
- Circuit Switched Data Call/Basic call (ETS 300 171/172)
- Calling/Connected Line Identity Presentation (ETS 300 173)
- Calling/Connected Name Presentation (ETS 300 237/238)
- Message Waiting Indication (ETS 301 260/255)

© 2008 AVAYA All rights reserved.

Messaging Networking

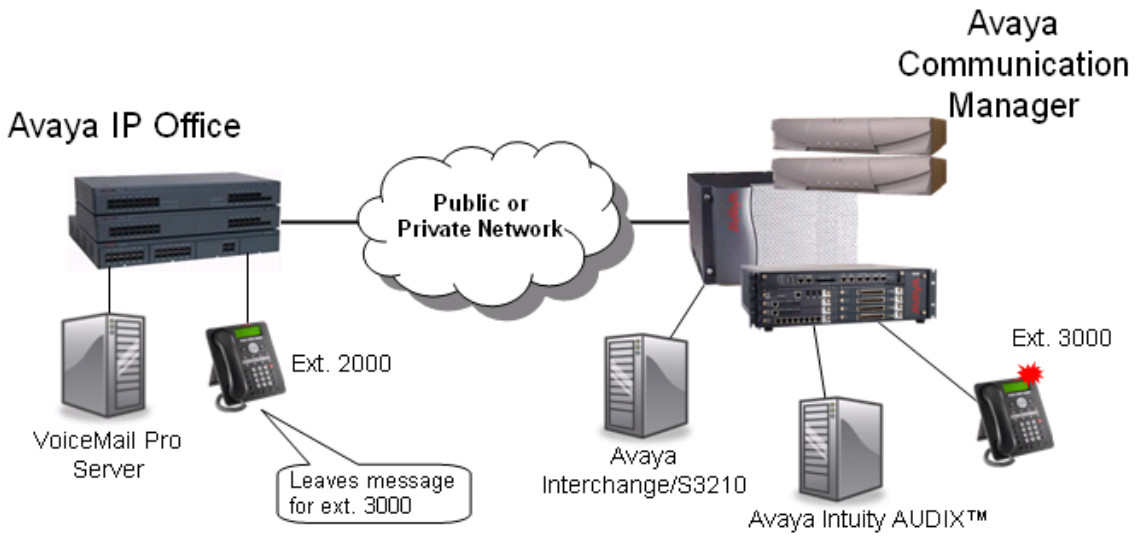
There are 2 options available today to provide messaging interoperability between IP Office and Definity / MultiVantage / ACM. The first option provides Centralized Voicemail while the second allows Avaya voicemail systems to be networked. The requirements, functionality and restrictions are summarized below:

- IP Office to DEFINITY / MultiVantage / ACM connected to Intuity AUDIX™ over a QSIG link (E1/T1 or IP)



- No local Voicemail required on remote branch IP Office but AUDIX RFA required on every IP Office
- Requires Intuity Audix 4.4+ connected via C-LAN to DEFINITY 9.5+ (see IP Office Offer Announcement dated August 2003 for more information on compatibility)
- Maximum of 19 IP Offices can be supported on 1 INTUITY AUDIX™ server (20 total with DEFINITY/ACM occupying one slot)
- Requires QSIG and Private Networking licenses on DEFINITY / MultiVantage / ACM
- User mailbox with Message Waiting Light support
- Forward voicemails between users
- No auto attendant (enhancement currently being investigated)
- No call recording
- No queuing at remote sites
- No Fax over IP to AUDIX™
- No Small Community Networking support when AUDIX™ is configured on IP Office.

- Avaya IP Office VoiceMail Pro networked to Avaya Modular Messaging / Octel / Intuity AUDIX™ via Interchange / S3210



- Requires local VoiceMail Pro on every branch IP Office licensed with VoiceMail Pro Networked Messaging RFA
- Requires Avaya Interchange/S3210 on Modular Messaging, Octel or Intuity Audix

© 2008 AVAYA All rights reserved.

- Provides 2,000 remote mailboxes per VoiceMail Pro server i.e. per branch office (to be extended to 10,000 remote mailboxes by next VoiceMail Pro maintenance release)
- User mailbox with Message Waiting Light support
- Forward voicemails between known remote users
- Fully-featured VoiceMail Pro at every branch office
- VoiceMail Pro Networked Messaging will only accept an incoming voicemail message for a local mailbox. It will NOT forward it to a remote Voicemail server. If required, this facility is available through Avaya Interchange.
- VoiceMail Pro Networked Messaging RFA is currently in extended trial and limited to Avaya Messaging Servers (not third-party messaging platforms)

© 2008 AVAYA All rights reserved.

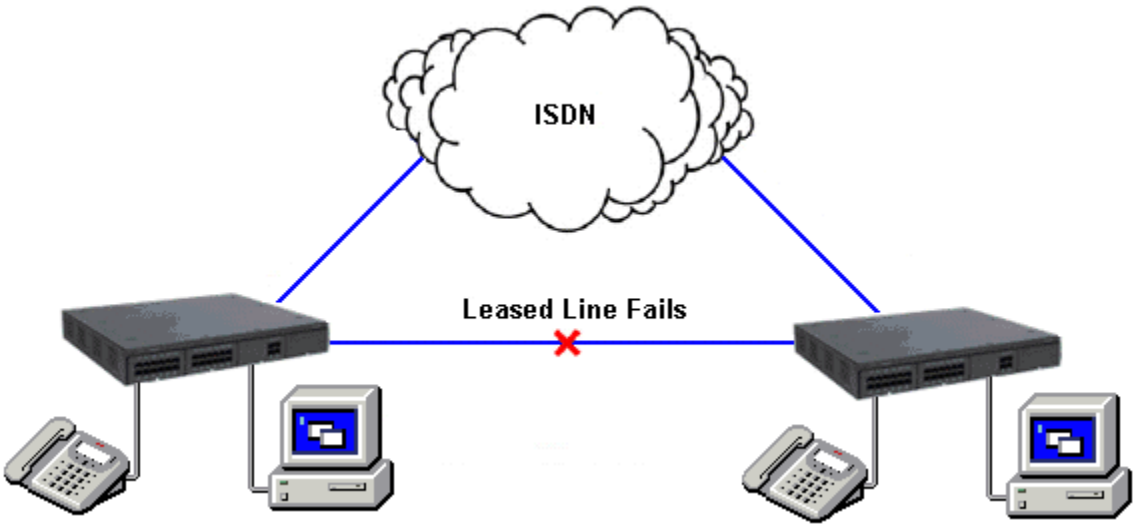
Alternate Route Selection

Alternate Route Selection (ARS) allows calls to be routed via the optimum carrier. Time profiles can also be used to allow customers to take advantage of cheaper rates or better quality at specific times of day.

If a primary trunk is unavailable or congested, then ARS provides automatic fallback to an available trunk (e. g., analog trunk fallback if a T1 or SIP trunk fails, or use PSTN for SCN fallback).

Multiple carriers are supported. For example, local calls are to go through one carrier between specific hours and international calls through an alternative carrier. Carrier selection using 2-stage call set up via in-band DTMF is possible. It is possible to assign specific routes on a per user basis, e.g. only allow expensive routes to be used by critical staff.

Note: Existing Least Cost Routing (LCR) configurations are automatically converted to ARS when upgrading to 4.1

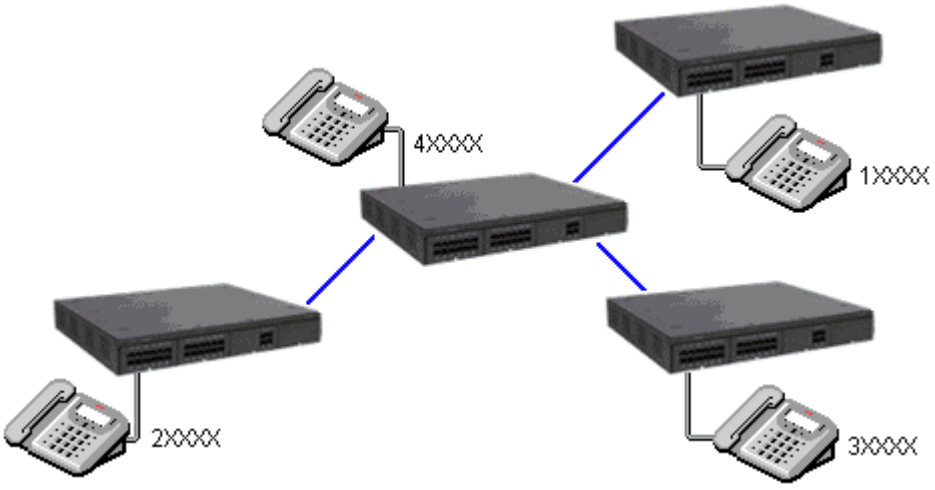


© 2008 AVAYA All rights reserved.

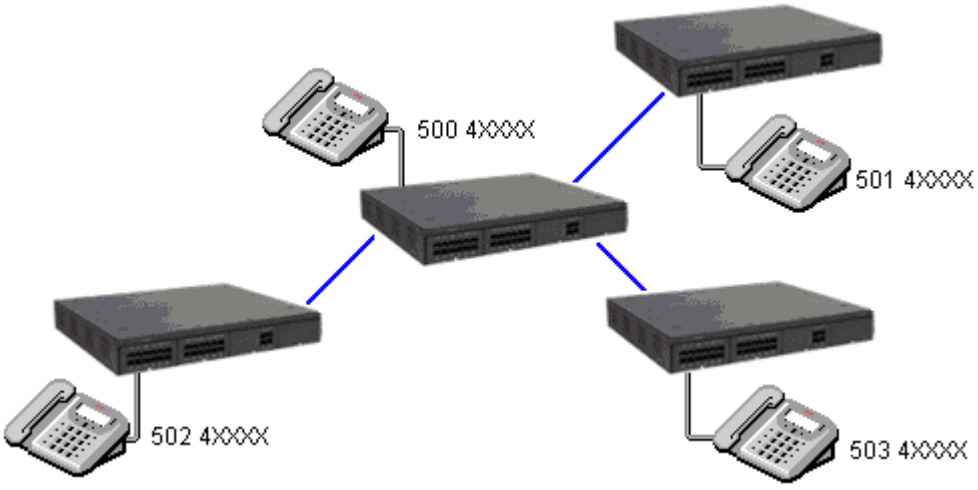
Network Numbering Schemes

IP Office uses fully flexible network numbering options. Dialed digits can be manipulated to add or remove digits, access codes etc. in order to fit into any numbering scheme. Two types of numbering schemes are commonly deployed - 'Linked Numbering' and 'Node Numbering' schemes. In linked numbering schemes each site within the network has a unique range of extension numbers and users simply dial the extension number of the called party. Often, linked numbering schemes are used in very small networks (< 5 sites) with less than 500 extensions. With node numbering schemes each site is given a node ID and this is prefixed by the user when dialing extensions at other sites. In this way extension numbers can be replicated across sites while still appearing unique across the network. Node numbering schemes are common in larger networks. Linked numbering schemes and node numbering schemes are sometimes both used within the same network with node numbering used at the large offices and linked numbering employed at clusters of satellite offices.

The following figures depict these two types of numbering schemes.



Linked Numbering Scheme



Node Numbering Scheme

© 2008 AVAYA All rights reserved.

© 2008 AVAYA All rights reserved.

Chapter 7.
Data Networking Services

© 2008 AVAYA All rights reserved.

7. Data Networking Services

LAN/WAN Services

Computers connected to an IP network in an office communicate via the LAN (Local Area Network). To support small networks both Small Office Edition and IP406 incorporate a Layer 2 Ethernet switch. The Small Office Edition supports 4 ports (with a fifth Ethernet port as a firewalled Layer 3 switch), the IP406 supports 8 ports. The IP412 and IP500 support a firewalled 2 port Layer 3 Ethernet Switch only.

When computers on the LAN communicate they do not care where the destination is, they just send messages with the address of the destination. These messages are likely to be received at all other computers on the same network but only one – the target destination – will act on the message. Where the destination is on another network, the router is needed to be the "gateway" to the rest of the world and find the optimum route to send the message on to the destination. The router alleviates the need to establish and hold a call for the duration of a communication session (when messages or IP packets are being sent between source and destination) by automatically establishing a connection only when data is to be passed. Routers may be connected together using WAN (Wide Area Network) links that could be point-to-point leased lines, managed IP networks, Frame Relay networks or exchange lines (Central Office). The IP Office system supports all of these types of network connections.

IP400 systems have a Wide Area Network (WAN) port that can be connected to a digital leased line service using either X.21 or V.35 interface at speeds up to 2048kbps. Point-to-Point protocol (PPP) is used over this link. The data within the call uses the Point-to-Point Protocol (PPP) which is used by the vast majority of manufacturers for linking routers. PPP support is essential if it is not the same manufacturer's equipment at each end of the link. Addition WAN ports can be added using up to two WAN3 10/100 external expansion modules. On Small Office Edition systems a single option WAN port card can be added. IP500 systems do not support this type of port.

Exchange lines (Central Office) can also be used in the event of failure of the WAN link or to provide alternate or top up bandwidth on demand.

All IP Office systems have an integral router with support for bandwidth on demand that allows the negotiation of extra bandwidth dynamically over time. Where connection is over ISDN, IP Office initiates extra data connections between sites only when there is data to be sent or sufficient data to warrant additional channels. It then drops the extra channels when they are no longer needed. The calls are made automatically, without the users being aware of when calls begin or end. The rules for making calls, how long to keep calls up etc, are configurable within IP Office.

It is possible to have several different routing destinations or paths active at any time linking the office to other offices and the Internet simultaneously.

Quality of Service

IP Office supports 802.1p packet prioritization. 802.1p is a specification for giving Layer 2 switches the ability to prioritize traffic (and perform dynamic multicast filtering). The prioritization specification works at the media access control (MAC) framing layer of the OSI model. To be compliant with 802.1p, Layer 2 switches must be capable of grouping incoming LAN packets into separate traffic classes. Eight classes are defined by 802.1p. Although network managers must determine actual mappings, IEEE has made broad recommendations. The highest priority is seven, which might go to network-critical traffic such as Routing Information Protocol and Open Shortest Path First table updates. Values five and six might be for delay-sensitive applications such as interactive video and voice. Data classes four through one range from controlled-load applications such as streaming multimedia and business-critical traffic - carrying SAP data, for instance - down to "loss eligible" traffic. The zero value is used as a best-effort default, invoked automatically when no other value has been set. In operation, 802.1p calls for the use of priority fields within the packet to signal the switch of the priority-handling requirements.

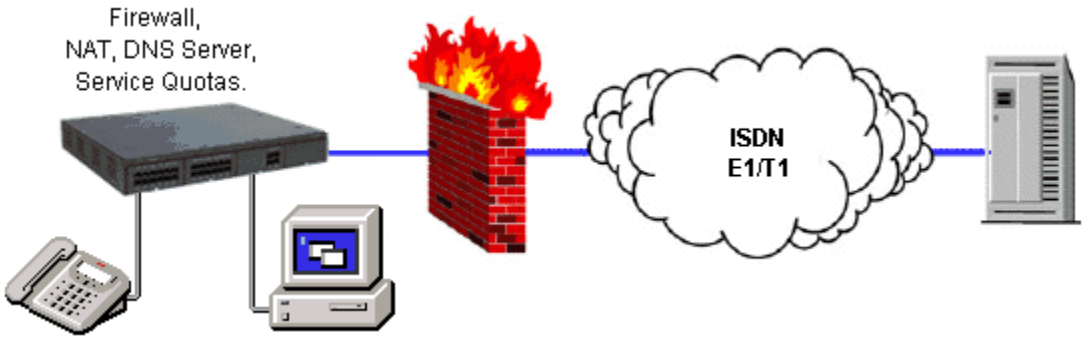
© 2008 AVAYA All rights reserved.

Internet Access

While the telephone is still the number one business communication tool, Internet access is becoming increasingly important for business-to-business communications. The ability to send and receive email, is now considered mandatory when dealing with many suppliers and customers, while access to the Internet for e-commerce applications and information has become vital.

IP Office systems provide shared, secure, high-speed access to the Internet via exchange lines (Central Office), digital leased line or IP VPN services.

Internet security concerns are addressed through the provision of an integrated firewall so removing the need for a standalone firewall. The firewall can be configured to cater for a variety of situations and will allow customers to control who can access external resources and when. The firewall isolates your private networks from the Internet, thereby ensuring that your network remains beyond the reach of hackers, while configurable service quotas can be set against a remote access service to ensure authorized users can gain access. Service Quotas place a time limit on outgoing calls to a particular IP Service so limiting costs. Each service can be configured with an alternative fall back, for example, you may wish to connect to your ISP during working hours and at other times take advantage of varying call charges from an alternative ISP. You could, therefore, set up one service to connect during peak times and another to act as fallback during the cheaper period.



Internet Access

© 2008 AVAYA All rights reserved.

Remote Access Features

IP Office's integral firewall, service quotas and timebands all apply to remote access calls. Remote access security can be supplemented by CHAP (encrypted passwords) to verify the end users, or PAP which does not support encryption. Timebands can control the hours within which the remote access service is available.

A "trusted location" can be set for dial in. These are locations that the System will allow either data access, e. g. a user dialing in from home, or access to voicemail without a voicemail code for a user collecting their voicemail messages from a mobile. The trusted location is also the location the Voicemail Server will call to inform the user of a new message.

Conversely a "specified location" can be set which restricts remote access from only that location, this specified location can also be a designated dial back number thereby minimizing the threat of unauthorized remote access.

IP Office systems can also incorporate remote access dial back services so that if a user always remotely accesses the office from a single location e.g. their home, then after logon verification the system will disconnect their call and dial them back. In addition to the added level of security dial back provides it can also be an excellent method of consolidating remote access charges onto the central office telephone bill.

In addition to remote access from Telephone Adaptors, an optional V.90 56Kbps modem module can be added to provide dial-in/dial-out to/from users equipped with analog modems. Also as standard, all ATM4 trunk cards and Small Office Editions analog trunk ports support switching of the first analog trunk to an integral V.32 modem for remote access.

LAN to LAN Routing

All businesses now have a need for data routing whether it's a requirement to share resources such as email servers, file servers and internet gateways, or seamlessly transport data between sites or network to and from their customers and suppliers. This is why each IP Office platform offers IP routing as standard.

Embedding a router within IP Office removes the costs, complexity and additional points of failure of external WAN multiplexers by allowing data and voice traffic to converge and share the network resources of IP Office. These network resources can range from dial up ISDN connections, point-to-point leased circuits, managed IP networks or Frame Relay as IP Office supports all these types of network connections.

© 2008 AVAYA All rights reserved.

Data Networking Features

Integral 10/100 Mbit Layer 2 Ethernet Switch

- **IP Office - Small Office Edition and IP406 V2 Only.**

All the IP Office - Small Office Edition platforms provide a four port Layer 2 Ethernet Switch. The IP406 V2 provides an 8 port Layer 2 Ethernet switch.

Each port auto-senses its operational speed, 10Mbps or 100Mbps. In addition to the four port layer 2 switch, IP Office - Small Office Edition has a fifth Ethernet port (labeled WAN) with its own IP Address (LAN2) intended for connecting to external xDSL or Cable Modems. This fifth port is a Layer 3 switch to the other four ports.

Integral 10/100 Mbit Layer 3 Ethernet Switch

Layer 3 switching is particularly useful in situations where it is desirable to have a 'trusted' and 'unsecured' network, where the 'unsecured' network is uncontrolled and carries public traffic on it.

It is possible to set up a firewall between two LAN segments using the IP Office layer 3 switch. Small Office Edition offers a firewall between its four port Layer 2 Ethernet switch and its Layer 3 Ethernet WAN port, while IP412 and IP500 support a two-port Layer 3 Ethernet switch with the firewall between them. Both of these switched ports have their own IP addresses (LAN1 and LAN2) and in order for traffic to pass from one port to the other, a route is configured in the system's routing tables.

From Release 4.1 onwards, port 8 of the IP406 V2 Ethernet Switch can optionally be configured as LAN2.

DHCP Server

IP Office can manage your IP Network for you through its integral DHCP Server. IP Office can be configured to hold a pool of IP addresses for users on the Local Area Network. When a user powers up their PC, the system will allocate them an IP address for the duration of their session. The DHCP server also provides the user's PC with the address of the Domain Name Service (DNS) server and the Windows Name Service (WINS) server. Alternatively, for customers who have a separate DHCP Server, IP Office can be configured to obtain its address from that DHCP server or be set with its own static IP address. Both IP Office - Small Office Edition and IP412 have two independent DHCP servers, one dedicated to each of the Layer 3 switched LANs.

Leased Line Support

All IP Office systems are capable of connecting to leased line services.

On IP400 systems X.21, V.35 and V.24 are supported via the WAN port on the back of the control unit. The X.21, V35 and V24 are externally clocked and can operate at any speed up to and including 2M. Additional WAN ports can be added using the WAN3 10/100 external expansion module. For the Small Office Edition an optional WAN port card can be added. This type of WAN connection is not supported on IP500 systems.

On all systems including IP500, WAN services are supported over E1/T1 PRI trunks and BRI trunks. E1/T1 trunks can be configured to operate in a fractional mode for 'point to multi-point' applications i.e. a single 2M interface could be treated as 3 x 512K and 8 x 64K going to 11 different locations. When using T1 as a Leased Line it is possible to use the same circuit for switched circuit services. Not all types of leased line are available in all territories, check for availability.

Dial-Up Circuit Support

Where the amount of traffic does not justify the cost of a dedicated leased line, the system can provide data connectivity via ISDN dial-up circuits using its E1/T1 or Basic Rate trunks. Where data speeds greater than a single channel are required (64K/56K), additional channels can be added to the call as and when they are needed.

© 2008 AVAYA All rights reserved.

Point-to-Point Protocol (PPP)

PPP is an industry standard Wide Area Networking Protocol, that allows inter-working with a wide range of 3rd party routers. PPP is used over dial-up or leased line circuits where a single channel is used to connect the two locations together. e.g. A single channel maybe a 64K channel on a dial-up circuit or a 256K leased line etc.

Multi-Link Point-to-Point Protocol (ML-PPP)

IP Office supports Multi-Link PPP allowing additional calls to be made where bandwidth greater than a single channel is required. The maximum number of channels available to data can be set on a service-by-service basis. When the available bandwidth reaches a user defined limit additional channels can be automatically added. Similarly, when traffic falls then the number of channels in use can be automatically reduced. If there is no data traffic on any of the channels in use then all lines can be cleared. Since most carriers have a minimum charge for calls, the period that a channel has to be idle before clearing is configurable. Through these mechanisms call costs can be effectively controlled while ensuring that bandwidth is available as and when it is needed.

Frame Relay

Frame relay is a wide area networking protocol based on the X.25 protocol. Individual network connections are multiplexed over a common medium by the use of Permanent Virtual Circuits (PVC). This allows a single Leased Line to provide connectivity to a number of different locations. Frame relay is currently implemented in IP Office as a CPE or 'router end' protocol over WAN connections. IP Office supports both PPP and RFC1490 encapsulation with fragmentation of large data packets to provide voice quality of service.

Service Quotas

IP Office can be configured to limit the maximum number of minutes that a service, such as Internet Access, is available for each user. This is the sum total of calls made and does not include periods of inactivity. Once the quota has been used the service is no longer available. The quota can be either automatically refreshed daily, weekly or monthly or manually refreshed by dialing a secure feature code on a handset.

Time Profiles

Time profiles set the operational times for service. For example, a time profile could be set up to make Internet Access available to staff only during lunch times. Using time profiles it is also possible to define an alternative service to operate outside the operational hours of the main service. This may be used to take advantage of alternative tariffs at off peak periods. Switching to this fallback service can also be controlled manually by dialing a secure short code from a handset. This can be particularly useful in allowing quick restoration of service in the event of an ISP failure. This feature also applies to days of the week or specific calendar dates.

© 2008 AVAYA All rights reserved.

Password Authentication Protocol (PAP)

PAP is a method of authenticating the remote end of a connection using unencrypted passwords.

Challenge Handshake Authentication Protocol (CHAP)

Challenge Handshake Authentication Protocol allows an incoming data call to be authenticated using encrypted passwords. The system also provides the option to periodically reaffirm the authenticity of the caller during the data call.

Data Header Compression

IP Header Compression (IPHC) reduces the header size of the data packet to gain bandwidth efficiency over Wide Area Networks, but adds to transmission latency.

Data Compression

IP Office supports both Microsoft Point to Point Compression and Stac Lemple Ziv to provide greater throughput on slow speed wide area network links.

Bandwidth Allocation Control Protocol (BACP)

Bandwidth Allocation Control Protocol allows the negotiation with the remote end of the data call to request additional calls to be made to improve aggregate data throughput.

Callback

Three types of call back are supported

- **LCP (Link Control Protocol)**
  After authentication the incoming call is dropped and an outgoing call is made to a predefined number to re-establish the link.
- **Callback CP (Microsoft's Callback Control Protocol)**
  After authentication from both ends, the incoming call is dropped and an outgoing call to a predefined number made to re-establish the link.
- **Extended CBCP (Extended Callback Control Protocol)**
  Similar to Callback CP however, the Microsoft application at the remote end will prompt for a telephone number. An outgoing call will then be made to that number to re-establish the link.

Domain Name Service (DNS) Proxy

Domain Name Service servers provide the translation of names such as www.avaya.com to the domain's IP address required to establish a connection. IP Office provides this service to PCs on the network by proxy.

© 2008 AVAYA All rights reserved.

Network Address Translation (NAT)

Network Address Translation is a mechanism that allows you to use different IP address on your private network behind a router with a public IP Address. When connecting to the Internet, ISPs typically want a customer to use an IP address they have allocated. Using NAT this is easily accommodated, eradicating the need for the customer to change their network numbering scheme and providing additional security to the internal users as their address in hidden to the public.

Typically, a company maps its internal network addresses to a global external IP address and unmaps the global IP address on incoming packets back into internal IP addresses. This helps ensure security since each outgoing or incoming request must go through a translation process. This also offers the opportunity to qualify or authenticate the request or match it to a previous request. NAT also conserves the number of global IP addresses that a company needs.

Proxy Address Resolution Protocol (ARP)

Support for Proxy Address Resolution Protocol allows IP Office to respond on behalf of the IP address of a device connected to it when receiving an ARP request.

Auto Connect

If a service is idle, that is no one is using the Internet, Auto Connect allows the IP Office to periodically connect to a service. This is ideal for mail polling to retrieve email from an Internet Service Provider. An 'Auto Connect Time Profile' controls the time period during which automatic calls are made, for example not at weekends or during the middle of the night.

Firewall

IP Office integrated firewall provides packet filtering of the most common IP protocols including File Transfer Protocol (FTP) and Internet browsing (HTTP). Each protocol passing through the firewall can be restricted/ allowed access in four different ways:

- **Drop**
  No sessions via this protocol will be allowed through the wall
- **In**
  An incoming session can "punch a hole" in the wall to allow traffic in both directions
- **Out**
  An outgoing session can "punch a hole" in the wall to allow traffic in both directions
- **Bothway**
  An incoming or outgoing sessions can "punch a hole" in the wall to allow traffic in both directions.

In cases where a protocol is not supported by default, the firewall can be customized to control packets based on their content.

IP Office allows the configuration of as many firewalls as needed through IP Office Manager. This permits different security regulations to be applied to individual dial-in users and data services.

© 2008 AVAYA All rights reserved.

Light-Weight Directory Access Protocol (LDAP)

IP Office supports LDAP directory synchronization. This allows the telephone number Directory (names and telephone numbers) held in IP Office to be synchronized with the information on an LDAP server (limited to 500 entries). Although targeted for interoperation with 'Windows 2000 Server Active Directory', the feature is sufficiently configurable to interoperate with any server that supports LDAP version 2 or higher.

Remote Access Server (RAS)

IP Office provides RAS functionality allowing external users to dial in to the local area network from modems, telephone adaptors and routers. Several of the previously described features and services can be applied to the dial-in users to create a powerful Remote Access Server. Dial-in users can be authenticated using either PAP or CHAP. Once authenticated the DHCP server can automatically assign the user an IP address to use while connected to the LAN. Individual time profiles and firewalls can be applied to the user restricting what they have access to and when they have access. For further security and accounting ease, IP Office can automatically call a user back. This keeps the cost of the telephone call on the company telephone bill removing the need to process individual expense claims.

Transaction Packet Assembler Dissembler (TPAD)

TPAD is a lightweight version of the X.25 protocol used in the retail market for transaction processing. Through faster transaction processing a retailer can reduce the floor limit of credit authorizations and benefit from lower transaction charges. A PDQ or credit card "swipe" telephone can utilize the digital trunks, via the DTE port or the USB on the rear of the IP Office. Since the link between the main unit and the transaction authenticator is digital no modems are required at either end.

Routing Information Protocol (RIP)

RIP is a distance vector protocol that allows routers to determine the shortest route to a destination network. It does this by measuring the number of intermediary routers that need to be traversed to reach the destination network. If more that one route exists to the same destination the shortest route is used. If a fault occurs on the shortest route it will be remarked as being infinite and any alternative route will become the new shortest route. This behavior can be used to add resilience into a data network. Where a customer has an existing data network comprising of third party routers, IP Office added to the network can provide back up using its routing and dial-up capability. RIP enabled routers share their knowledge of the network with each other by advertising and listening to routing table changes. IP Office Supports both the RIP I and RIP II standards.

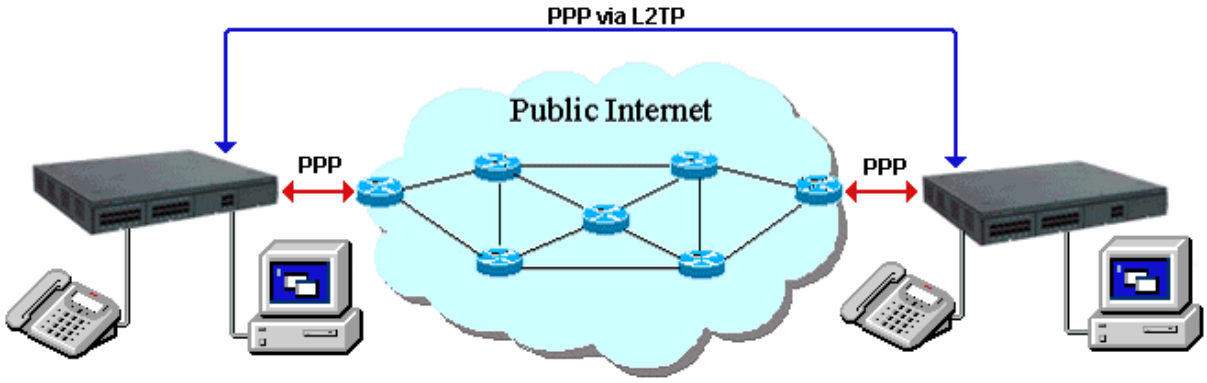
© 2008 AVAYA All rights reserved.

VPN: IPSec Tunneling

IPSec tunnels allow a company to pass data between locations over unsecured IP networks such as the public internet. The company data is secured using 3DES encryption making it unintelligible to other parties that might be 'eaves dropping' on the traffic. Tunneling can be applied to link offices together or provide workers access to the office over the internet. All IP Office systems support up to a total of 256K worth of encrypted traffic to multiple locations. Initially, inter-working is supported only between IP Offices that are connected either directly on a WAN port or via the LAN using a 3rd Party router. IPSec is optional and enabled on IP Office through a License Key.

Note: Check with Avaya for supported scenarios and 3rd party devices.

VPN: Layer 2 Tunneling Protocol



PPP authentication using PAP or CHAP takes place between directly connected routers only. When using a public IP Network to connect sites this authentication takes place between the customers router and the service provide router that it is connected to. In some circumstances it is desirable to authenticate between the customer owned routers, jumping over all the intermediary routers of the service provide network. Layer 2 Tunneling Protocol allow this to happen by facilitating a two stage authentication, firstly with the service provider router then the customer router on the remote network.

© 2008 AVAYA All rights reserved.

© 2008 AVAYA All rights reserved.

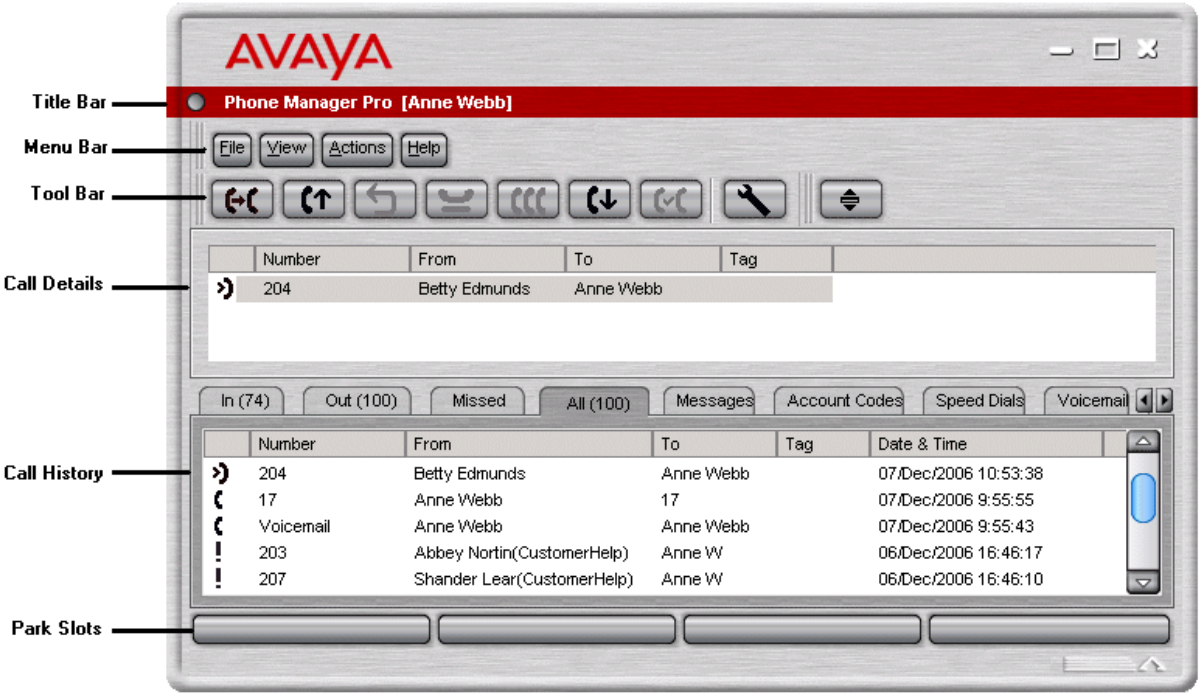
Chapter 8.
Phone Manager

© 2008 AVAYA All rights reserved.

8. Phone Manager

The IP Office Phone Manager application provides users control of their telephone from a networked PC.

Phone Manager can be used with any IP Office extension; analog, digital or any IP telephones, wired or wireless, and is available in three versions: Phone Manager Lite, Phone Manager Pro and Phone Manager PC Softphone subject to licensing.

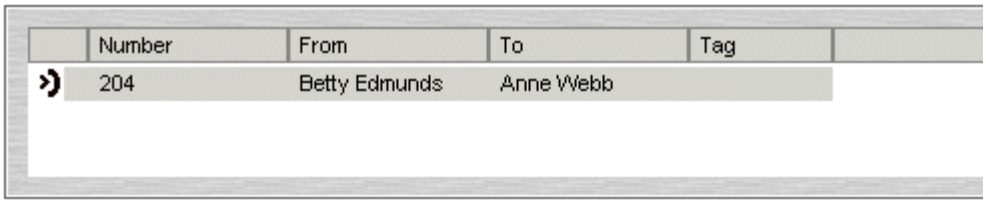


© 2008 AVAYA All rights reserved.

Phone Manager Lite

Phone Manager Lite is included as part of the IP Office solution free of charge for every user and provides easy access to telephony features, call information and call control. Phone Manager will normally run in the Windows system tray once the user is logged on, minimizing screen space when not in use.

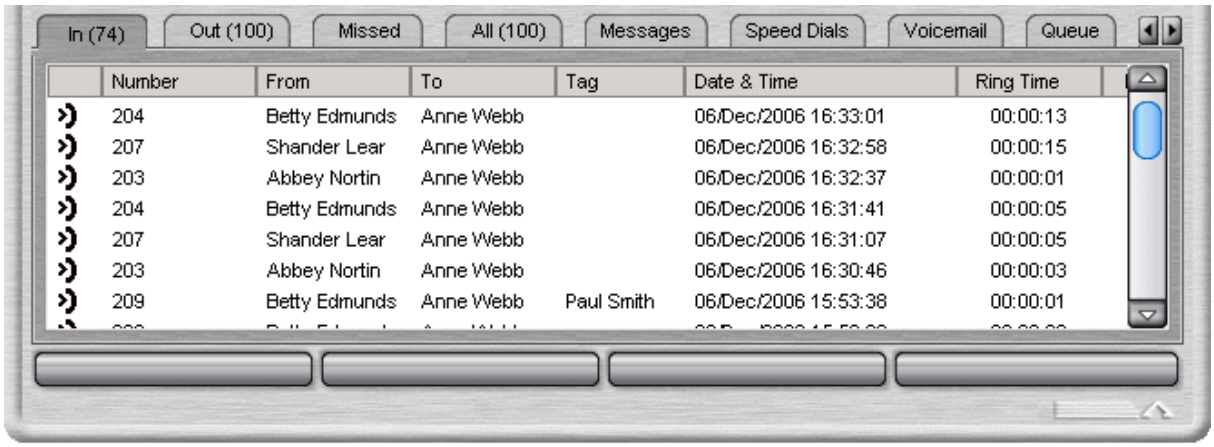
Caller ID/Name Presentation



Caller ID is presented as standard (where provided) allowing users to see who's calling before answering. The caller's phone number and name (if known to IP Office) are clearly shown in the call status area of the Phone Manager screen. For incoming calls, the dialed destination is also visible, for example the user's Direct Dial number, or a specific department, for example switchboard, sales, support or administration.

This feature allows users to answer the call appropriately and gives the flexibility to participate in multiple hunt groups, particularly important for small businesses. The same information is also displayed should a second incoming call be presented, allowing users to easily switch between calls or allow the second call to go to voicemail. Users can choose to have the information pop-up on their PC automatically as soon as a call is presented, or when the call is answered.

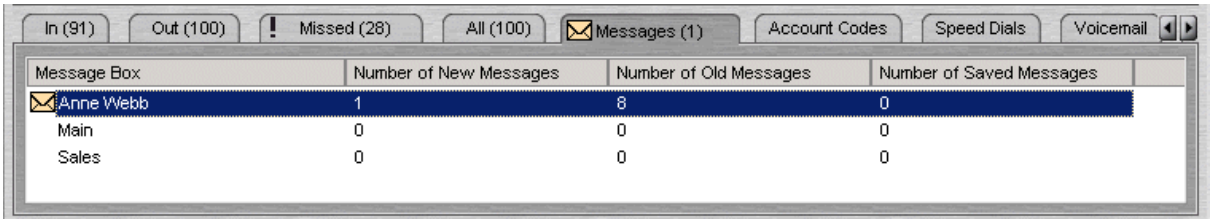
Call History



Phone Manager's call history keeps a combined record of up to 100 calls while the application is active. Double-clicking any logged call dials that number. If Advice Of Charge service is available from the ISDN service provider, this will also be displayed for outgoing calls.

© 2008 AVAYA All rights reserved.

Voicemail Access



Phone Manager Lite provides notification of any new voicemails received and provides access into the user or group's mailbox allowing messages to be played.

Desktop PC Telephony Controls

Phone Manager has telephony buttons on a tool bar that activate standard telephone functions such as Answer, Transfer, Hold, Account codes and Conference etc. so that users don't need to remember IP Office specific feature codes. Personal settings such as Do Not Disturb (including exceptions list), call forwarding, mobile twinning and voicemail transfer option settings can be easily set up using Phone Manager.

Calls can be easily parked using "drag and drop" functionality. Four Call Park slots/zones, which can be shared between users and operators, or within a department on the same IP Office system, further add to the ease with which the entire call handling process is streamlined with Phone Manager.

© 2008 AVAYA All rights reserved.

Personal Productivity and Collaboration

All versions of Phone Manager feature a Busy Lamp Field (BLF) and Speed Dials. This allows users to customize the application to reflect the status of their department, immediate colleagues or the whole company as desired. The Direct Station Select allows you to dial regularly used internal and external numbers via a single-click. A single Direct Station Select icon allows you to dial their work, mobile/cell phone and home numbers. The Busy Lamp Field feature allows you to see at a glance, who is available to take a call, who is already on a call and who has placed their phone on Do Not Disturb. BLF information is also available on remote users as long as they are on a Small Community Network (SCN). Phone Manager Lite supports up to 15 Speed-Dial/BLF entries.

Internal User		External Number	
	Busy		Work
	Message		Mobile
	Divert		Home
	Do Not Disturb		Fax
	Not logged into LCS		
	Logged into LCS		

Where Microsoft Live Communications Server (LCS) is also available within the user's business, Phone Manager users can view colleague's presence (online, offline) as well as send Instant Messages (IM) via Phone Manager. For example users can send an IM to alert a colleague that an important call is waiting for them even though they're busy on another call.

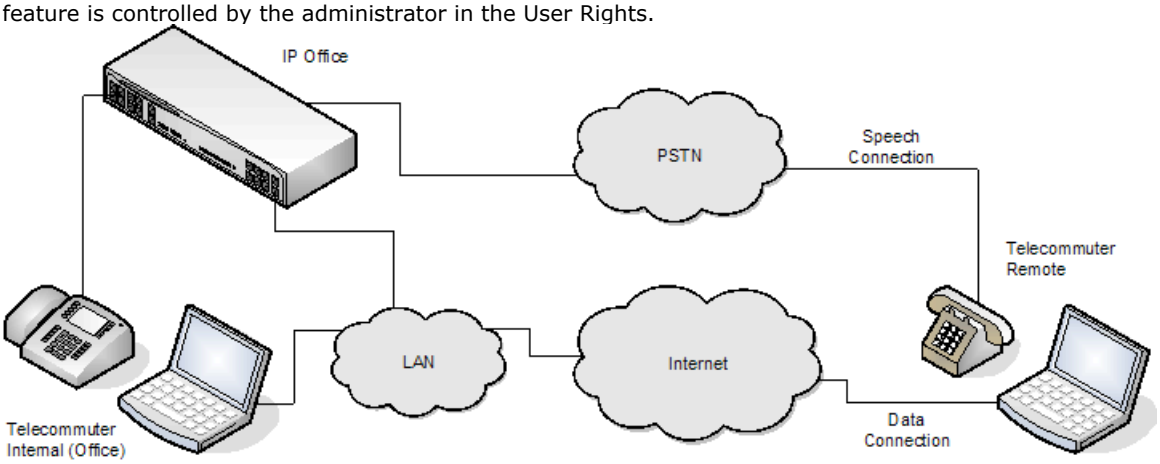
Phone Manager also offers Conferencing Center toolbar buttons that allow users to book a conference or join a web conference. Note: The booking feature is only available if the user has been granted permission by the system administrator and Conferencing Center has been installed (see the Conferencing Center section for further details).

© 2008 AVAYA All rights reserved.

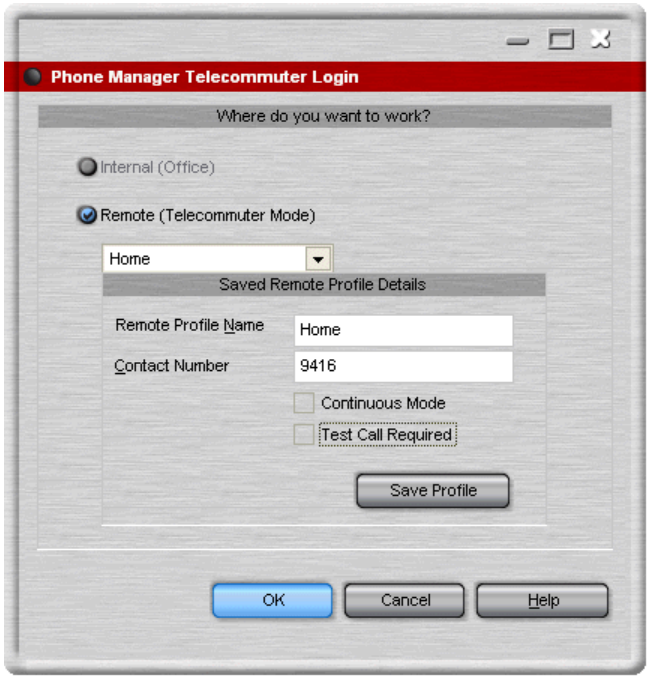
Phone Manager Pro

Phone Manager Pro is licensed on a per-user basis and provides all of the Phone Manager Lite features plus the following:

- **Personal Productivity and Collaboration**
  Phone Manager Pro offers increased capacity from 15 to 1000 Speed Dial/BLF entries. These are distributed across 10 tabs to allow users to group speed-dial/Busy Lamp Field icons by department or location, for example Sales, Support, etc. Each speed-dial tab supports up to 100 speed dial/BLF entries.
- **Telecommuter Mode**
  Phone Manager Pro allows making and receiving calls and retrieving voicemails on an external phone number as if they were in the office, with Phone Manager providing the call control. It also provides billing convenience and potential cost savings for remote workers and mobile work force. Access to the feature is controlled by the administrator in the User Rights.



When logging on, a Telecommuter user will be asked to choose the phone number they can be reached at. This number is either entered directly or is part of a previously saved profile. Once selected, Phone Manager will use this number to make and receive calls and retrieve voicemails for the duration of the session.



© 2008 AVAYA All rights reserved.

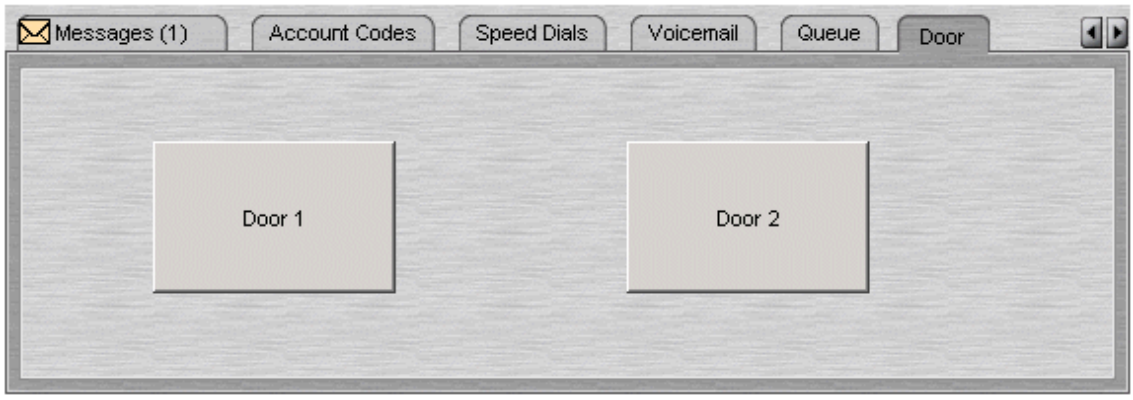
- **Integration with Contact Management packages**
  To facilitate screen popping of the contact details of an incoming caller, dialing from the contact record with a simple mouse click and simple creation of new contact records with auto-insertion of the telephone number while on a call. The user can select which Contact Management should be popped:
  - **Outlook**
  - **GoldMine**
  - **ACT!** *(*ACT! 7.0 and higher requires the TAPI.NET add-on from various providers plus the IP Office TAPI driver from Avaya.)*
  - **Maximizer.**
- **VoiceMail Pro mailbox control**
  - **Manage voicemails**
    Phone Manager Pro allows users to play, rewind, fast-forward, save or delete their voice messages.
  - **Manage Personal Distribution Lists**
    Phone Manager Pro allows users to configure their Personal Distribution Lists (VoiceMail Pro Intuity mode only).
  - **Manage voicemail greetings**
    Users can record and select which of the personal greetings is active (VoiceMail Pro Intuity mode only).
  - **Personal Directory**
    Personal phone number directory which allows further personalization and improves productivity:
- **Name matching**
  If the Caller ID is recognized in the local PC directory, the caller's name can be displayed. Up to 1000 entries are supported.
- **Simple incoming call scripting**
  Scripts can be displayed based on the Caller ID or the dialed number (DID/DDI) to remind users of a specific greeting or message to use.
- **Distinctive ringing**
  Allows the configuration of distinct ringing on a per caller basis. WAV sound files can be associated with incoming callers' numbers and then played through the PC speakers when a call is received from that number. This allows you to easily differentiate calls from important customers, clients, and unknown callers.
- **Compact Mode**
  Compact mode minimizes the screen space required to run the Phone Manager Pro application. While in compact mode, a notification slider alerts new calls and allows the user to view the caller ID or associated caller's name and answer the call. Users can easily switch between standard and compact modes.

- **Agent Mode**
  Agent mode operation allows the user to perform contact center functionality without needing a specially designed contact center telephone, for example one with dedicated keys such as log on/off. Agent-mode users can set their phone to "Busy" or "Wrap-Up" and select which hunt group they are member of via simple button clicks. Access to this feature is controlled by the administrator via User Rights.

- **Account Codes tab**
  Users can easily activate Account codes (before or during the call) through the 'Account Codes' tab. This tags calls with an alphanumeric account code via a single-click. Note: Lite users can enter account codes but cannot view the Account Codes tab.
- **Queue monitoring**
  Queue monitoring allows the user to see the number of calls waiting in up to 2 queues. The Phone Manager Pro user does not need to be part of the hunt groups being monitored.

© 2008 AVAYA All rights reserved.

- **Door entry control**
  Door entry control allows the user to open or close the two external relays in the IP Office system. This can be used to activate an external system such as door-entry or security camera.



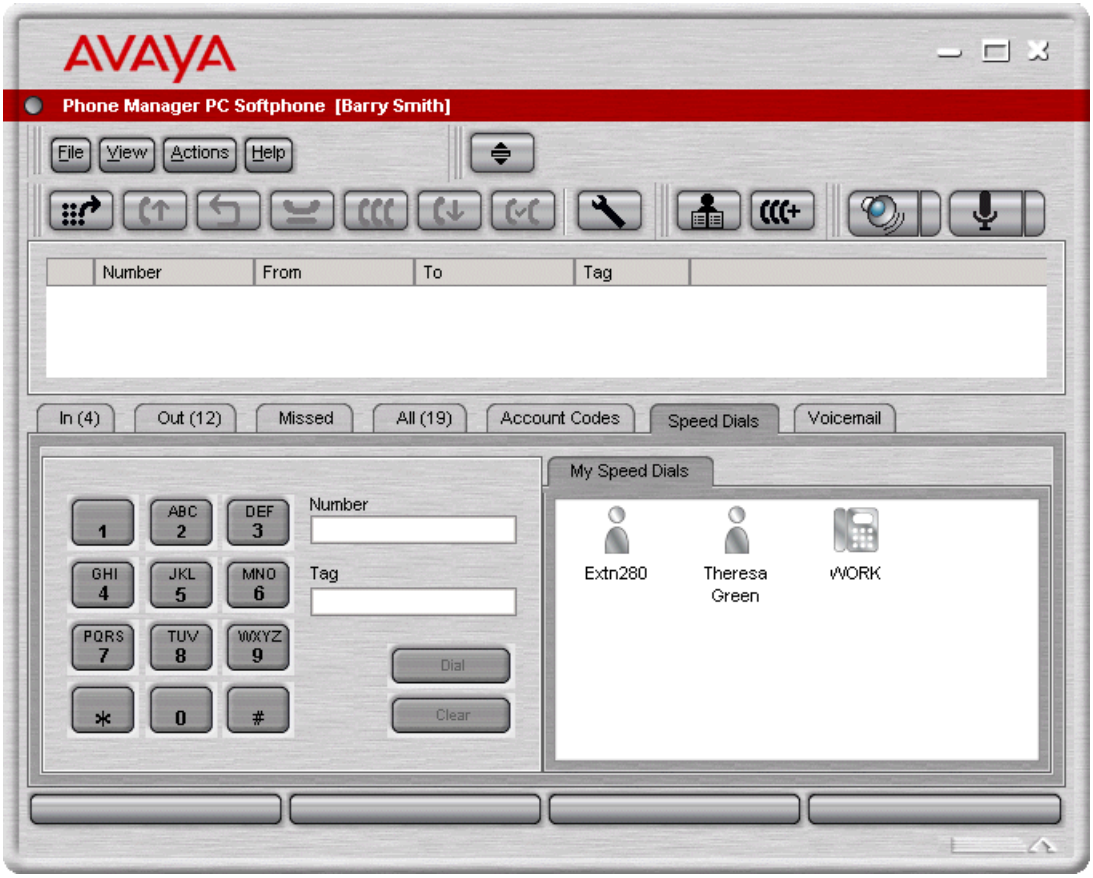
- **Call History**
  Phone Manager Pro provides separate tabs for Incoming, Outgoing, Missed and All Calls. Each call log tab will store the last 100 entries which can be sorted by date and time, caller ID and call duration if required.

© 2008 AVAYA All rights reserved.

Phone Manager PC Softphone (IP Softphone)

Phone Manager PC Softphone is licensed on a per-user basis and provides all of the Phone Manager Pro functionality. In PC Softphone mode, both audio and call control operations are handled on the PC so no additional telephone is needed. When using PC Softphone, the user will need an audio device such as a headset or USB handset, both USB and soundcard interfaces can be used with PC Softphone.

PC Softphone can be twinned with another IP Office extension offering mobility and choice so that calls can be answered on either endpoint.



Phone Manager PC Softphone has the significant advantage for mobile users with wireless or wired remote access to the LAN, providing 'a phone within their laptop' with all the features available in the office. From IP Office 4.2 onwards, users can also login as a call center agent using Softphone. PC Softphone is supported on Windows Vista Business/Ultimate.

© 2008 AVAYA All rights reserved.

Phone Manager Feature Summary

Feature	Phone Manager Lite	Phone Manager Pro and PC SoftPhone
Inbound/outbound call handling.	Yes	Yes
Phone call control.	Yes	Yes
Configure phone preferences.	Yes	Yes
Configure keyboard short cuts.	Yes	Yes
CLI (ANI) / Name display.	Yes	Yes
Speed dial / Busy Lamp Field management.	Yes - 15 icons maximum.	Yes - 100 icons maximum per tab.
Speed Dial tabs (to group Busy Lamp Field icons)	Yes - 1 tab.	Yes - 10 tabs maximum.
Microsoft Live Communications Server (LCS) Integration	Yes	Yes
View internal users' presence via LCS	Yes	Yes
Send Instant Messages (IM) to internal users via LCS	Yes	Yes
Telecommuter mode	–	Yes (not PC SoftPhone)
Compact mode	–	Yes
Local Phone Directory.	–	Yes - 1000 entries maximum.
Call history log – all, missed, messages.	Yes	Yes
Separated incoming/outgoing call log.	–	Yes
Collect new voicemail messages.	Yes	Yes
Voicemail box control (Intuity and IP Office modes).	–	Yes
Personal Distribution List set up (Intuity mode)	–	Yes
Incoming call scripting.	–	Yes
Time on call.	–	Yes
Advice of Charge (ISDN service provider dependent)	Yes	Yes
Door opening control.	–	Yes
Queue monitoring.	–	Yes - 2 Queues
Conference Control Display.	Yes	Yes
Conferencing Center action buttons	Yes	Yes
'Screen pop' contacts (Outlook, Goldmine, ACT! and Maximizer).	–	Yes
Simple Outlook contact record creation.	–	Yes
Agent Mode.	–	Yes
Distinctive Ringing (WAV file).	–	Yes
Post Connect dial (sending DTMF while connected to another party).	Yes	Yes
VoIP mode (to run as an PC Softphone)	–	Optional license

© 2008 AVAYA All rights reserved.

Phone Manager PC Requirements

- **Phone support:**

  Any telephones connected to IP Office, although hands-free operation is only supported on suitable Avaya Digital and IP telephones.

- **PC requirements:**

  - Always refer to the latest Avaya IP Office Technical Tip or Technical Bulletin for any updated information with regard to Operating Systems, Service Packs or PC hardware
  - Refer to Technical Specifications section of the Product Description for Operating System and Hardware requirements

- **Licensing:**

  - **Phone Manager Pro:**
    Requires a Phone Manager Pro license for each user.
  - **Phone Manager PC Softphone:**
    Requires an IP Office PC SoftPhone license in addition to the Phone Manager Pro user license. There must be equal or greater Phone Manager Pro licenses than PC SoftPhone licenses. The use of a headset is strongly recommended. Operation through standard speakers and integral PC microphones is possible but not recommended.
    - Phone Manager Pro screen popping provides integration with Microsoft Outlook 2000/2003/XP, Act! 6.0 and 2005, Maximizer 7.5 and 8.0 Enterprise, Goldmine 6.0 and 6.7.
    - Phone Manager PC Softphone supports QoS in the form of DiffServ for both Windows XP/2000.
    - Phone Manager PC Softphone can be used over a wireless LAN; in this configuration no more than 3 simultaneous calls can be supported per access point.
  - Up to 360 Phone Manager users can be supported on the same LAN subnet as IP Office. Where remote subnet Phone Manager users are deployed, up to 10 remote users will receive BLF updates.
  - Up to 5 Phone Manager users can be supported on a single Citrix thin client server
  - Instant Messaging options require the network to have a Microsoft Live Communication Server (LCS) with both a server license and client licenses for each user. Phone Manager has been verified as compatible with Microsoft LCS 2003 and 2005. No additional license is required in IP Office.

© 2008 AVAYA All rights reserved.

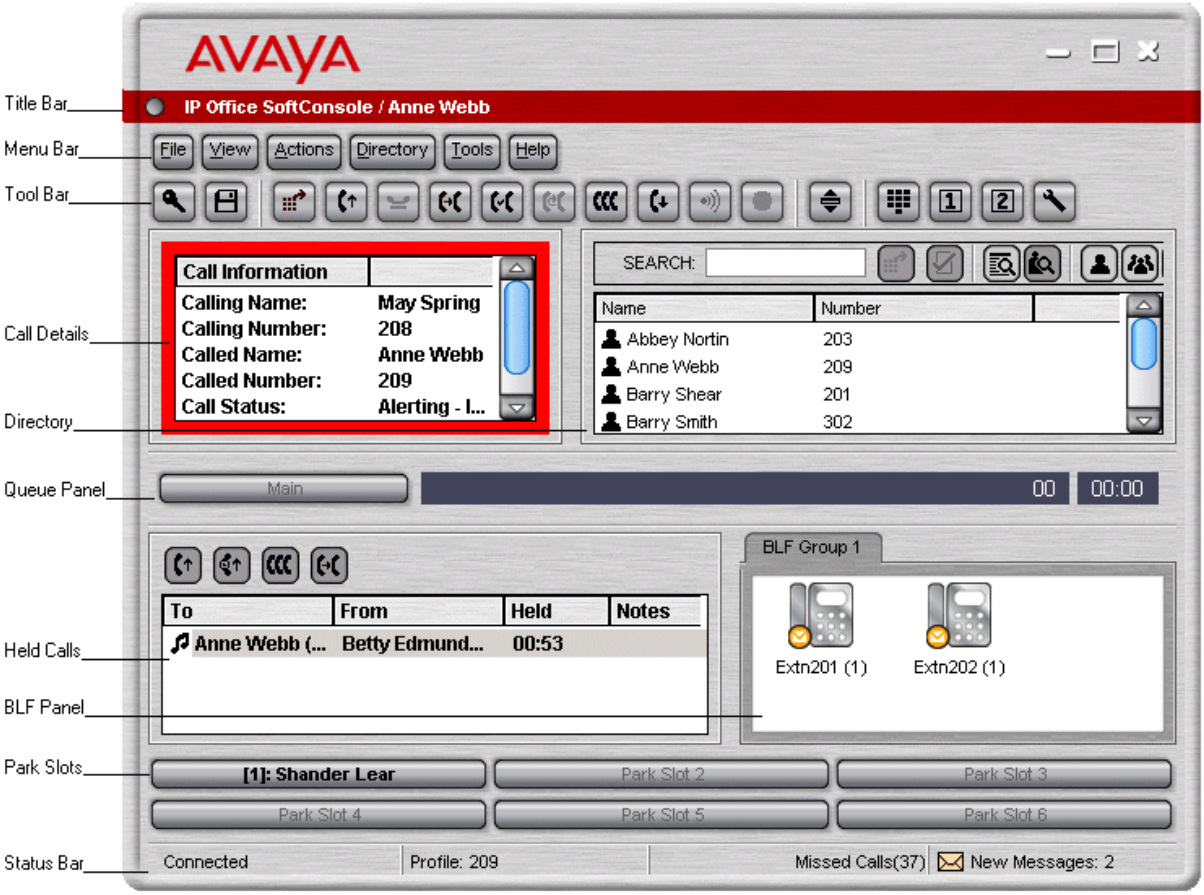
© 2008 AVAYA All rights reserved.

Chapter 9. SoftConsole

© 2008 AVAYA All rights reserved.

9. SoftConsole

SoftConsole is the PC based Windows Operator Console for IP Office. SoftConsole has been designed to improve operator service by providing the operator with call information and available call actions to simplify call handling and give the appropriate response to the caller. With this easy to use software tool the operator can maintain visibility of the number and type of calls waiting and so ensure that clients are greeted in a professional manner. SoftConsole has a similar look and feel to the Phone Manger application and can be minimized in the Windows system tray when not in use, but will pop up on the screen when a call is received.



SoftConsole has been designed to be easy to use, while offering a look and feel, which will appeal to experienced and novice operators alike.

© 2008 AVAYA All rights reserved.

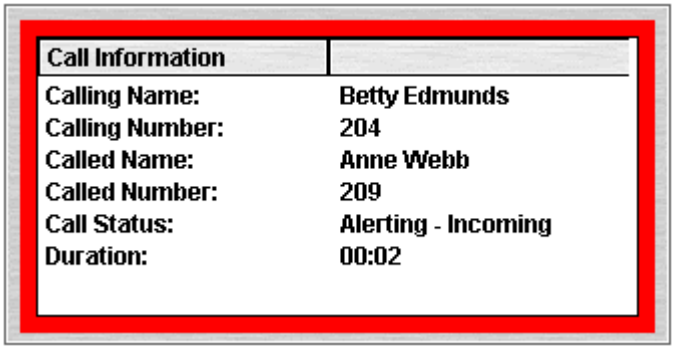
The SoftConsole screen is divided into the following areas:

- **Main Menu Bar**

Commands and actions are available through menus. Some features can only be used when the right conditions. If they are not available, the feature will be "grayed out" until conditions change that allow the feature to be used. The following features are available on the tool bar:

- Login.
- Save Profile.
- New call.
- Answer call.
- Hold call.
- Transfer call.
- Transfer complete.
- Reattempt transfer.
- Conference.
- Hang up.
- Page.
- Record call.
- Compact view.
- Dial Pad.
- Access conference room 1.
- Access conference room 2.
- Options.

- **Call Details Panel**



The call details panel on the left shows details of the current call which will include the following information:

- **Calling Name**
  The system directory name associated with the calling number.
- **Calling Number**
  The telephone number of the call originator.
- **Called Name**
  The system user name or hunt group name associated with the called number.
- **Called Number**
  The extension number the incoming call has been routed to by the system.
- **Call Status**
  States the progress of a call. The border around the call status panel changes color to indicate the status of the call.
- **Call Duration**
  The length of time that the has been in the state as indicated by the Call Status
- **Notes**
  This area displays notes or information about the call i.e. when a call has been returned as there was no answer from the extension it was transferred to. If annotation is attached to the call, details are shown in the Notes area.

If a new call arrives, the call details panel will display the calls waiting to alert the operator and allow answering of the call based on the Caller ID.

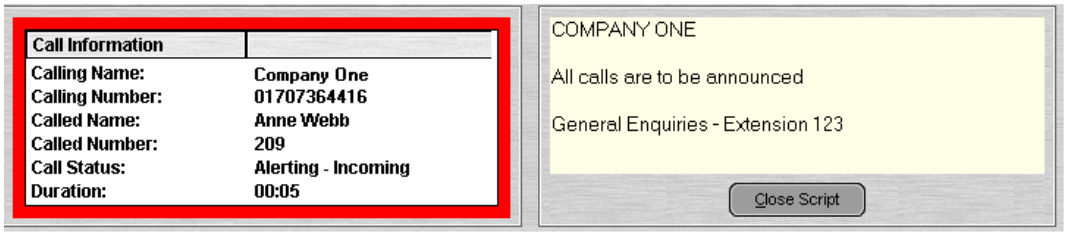
© 2008 AVAYA All rights reserved.

- **Directory Panel**

Property	Value
Name:	Anne Webb
Number:	209
Busy Status:	Idle
Do Not Disturb Status:	Off
Login Status:	Logged In
Group Status:	
Main	Out of Group
Sales	In Group
CustomerHelp	In Group
Absent Message:	
New Voice Mail Messages:	2
Forwarding Status:	
Forward Unconditional:	Off
Forward On No Answer:	Off
Forward On Busy:	Off
Follow Me:	Off
Forward Hunt Group Calls:	Off

The directory panel on the right shows information on following:

- **Directory entries**
  Including IP Office users, hunt groups and external directory user (non IP Office extensions)
- **Single directory entry details**
  Including IP Office users, Hunt Groups and external directory user (non IP Office user).
- **Script**
  When a script has been configured for either the calling or called number, the script is displayed in this panel. For example, an operator may be answering calls on behalf of more than one company. To ensure the call is answered with the correct company name a script file can be created with the company name details. The script is displayed whenever a call is received for that company.



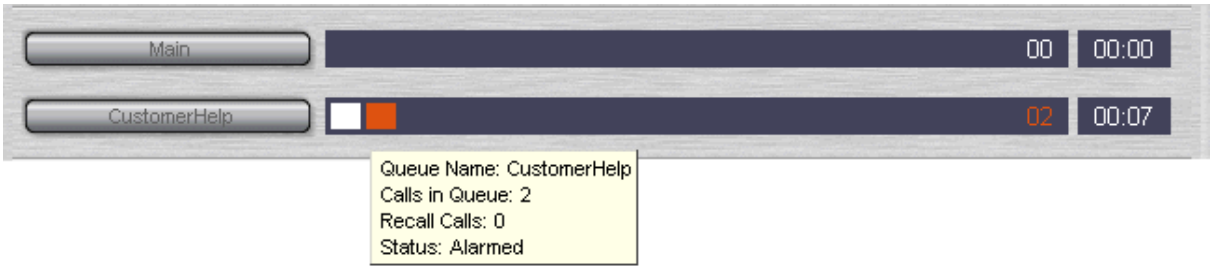
- **Conferencing**
  Within SoftConsole, calls can be conferenced when held, or a conference can be created through the two conference rooms:
  - **Conference Held Calls**
    An operator can conference calls that are in the Held Panel. All calls in the Held Panel will be conferenced.

© 2008 AVAYA All rights reserved.

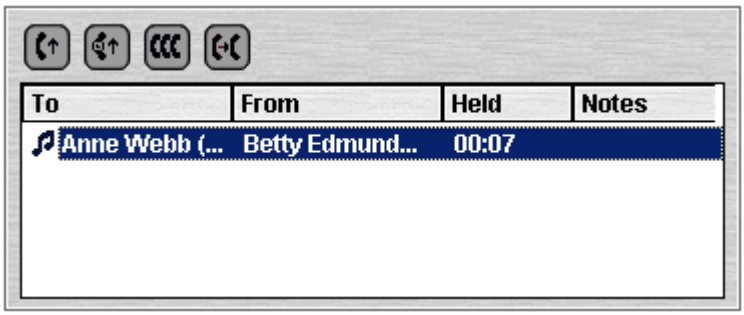
- **Conference Room**
  An operator can configure up to two conference rooms including details on who is hosting the conference plus the ability to send out invites to conference participants (automatic invites can be generated in conjunction with VoiceMail Pro, see IP Office Conferencing Center for more details). SoftConsole gives the operator visual status of calls in the conference room:

Not Invited.	Invited.	Joined.	Declined.	Unavailable.

- **Queue Panel**
  The queue panel displays a bar graph of the number and the status of external calls held in a particular queue. Up to 8 Call queues can be configured and labeled to reflect incoming calls for specific Hunt Groups.



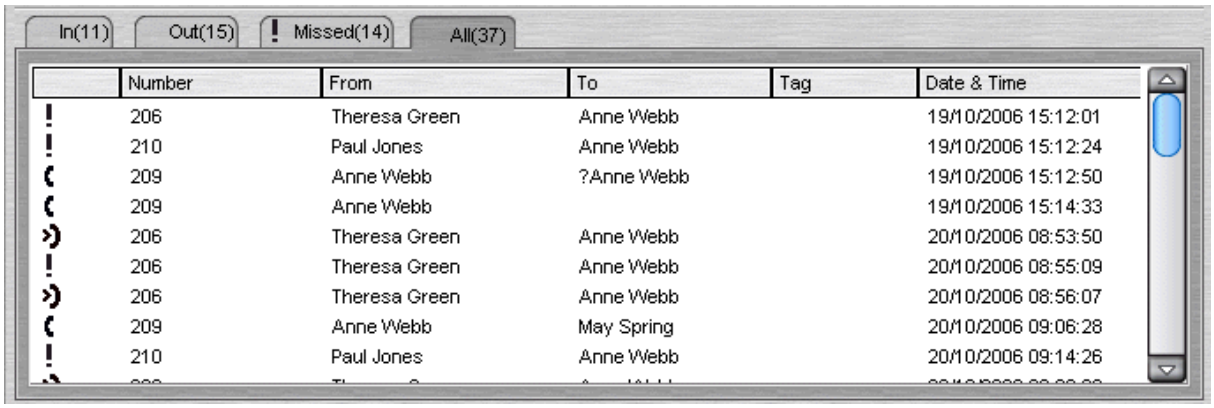
- **Held Calls Panel**
  The held call panel enables the operator to manage all calls held at the operator station. These calls will appear as a list in panel. The operator can perform the following the functions: Answer the highlighted held call, Answer the longest held call, Conference held calls (see conferencing section above) or Transfer held call.



- **BLF Panel (Busy Lamp Field Panel)**
  The BLF panel displays icons to indicate the status of selected users. Each icon provides information on individual users such as: Unread 'User' voicemail messages, User status information, for example Busy, DND and Forwarded is indicated by the various icons used. Up to 10 tabs with 100 icons on each tab are supported.

© 2008 AVAYA All rights reserved.

- **Park Slot Panel**
  The park slot panel can contain up to 16 system-wide park slots with specific Park ID's for each slot.
- **Call History**
  SoftConsole's call history keeps a combined record of up to 100 (incoming, outgoing and missed) calls while the application is active Double-clicking any logged call dials that number.


In(11)
Out(15)
Missed(14)
All(37)
Number
From
To
Tag
Date & Time
206 Theresa Green Anne Webb 19/10/2006 15:12:01
210 Paul Jones Anne Webb 19/10/2006 15:12:24
209 Anne Webb ?Anne Webb 19/10/2006 15:12:50
209 Anne Webb 19/10/2006 15:14:33
206 Theresa Green Anne Webb 20/10/2006 08:53:50
206 Theresa Green Anne Webb 20/10/2006 08:55:09
206 Theresa Green Anne Webb 20/10/2006 08:56:07
209 Anne Webb May Spring 20/10/2006 09:06:28
210 Paul Jones Anne Webb 20/10/2006 09:14:26

- **Status Bar**
  This Shows current status of the system and is divided into four sections that display current connection status, current Profile name, information messages and The number of new voice mail messages for the operator. Information messages include any alarm conditions that are present within the system.

© 2008 AVAYA All rights reserved.

SoftConsole Options

SoftConsole has many configurable options available to the operator to personalize the look and feel. The Operator can tailor the usability specifically to each their personal preferences. The following configuration options are available:

- **Incoming Calls**
  This tab enables the operator to manage the local SoftConsole directory by creating, editing and deleting entries from the selected directory. Also the operator is able to associate a script or media file with each specific entry.
- **Queue Mode**
  This tab enables the operator to configure the queue window with up to 8 hunt group queues, which will include a recall queue. Queues can be created, edited and deleted while also providing the operator with the additional benefit of positioning them in the queue window in order of operator preference. Management by exception is used to monitor queue status by enabling the operator to set up various alarm thresholds such as the Number of calls in queue and Longest waiting call time. A WAV media file can be associated with an alarm for further customization.
- **Park Slots**
  This tab enables the operator to define which park slots are accessible on a system wide basis up to a maximum of 16. The operator is also able to assign which numbers are used to access each park slot and where the slot appears in the park slot panel.
- **BLF Groups**
  This tab allows the operator to create and edit BLF groups.
- **Door Entry**
  This tab allows the operator to configure up to two door entries.
- **Directories**
  This tab enables the operator to choose access to the following directories: SoftConsole local directory, IP Office system directory and Microsoft Outlook contacts. Once chosen, the operator is able to map fields to directory entries.
- **Conferencing**
  This tab allows the operator to set up the names of the two conference rooms. The names will appear on the telephone displays of users in the conference room (maximum of 10 characters).
- **Keyboard Mapping**
  This tab allows the operator to assign keyboard short cut keys for SoftConsole functions.
- **Keyboard Actions**
  This tab allows the operator to specify the default action when alphabetic or numeric characters are pressed.
  - **Alphabetic Keystrokes:** Begin directory search or Open call annotation window
  - **Numeric Keystrokes:** Begin directory search or Open pop-up dial pad
- **Appearance**
  This tab allows the operator to change the appearance of SoftConsole fonts, skins and the call information window color.
- **SoftConsole**
  This tab allows the operator to save the changes made to the configuration of SoftConsole either automatically or manually to a local configuration file on the PC.

© 2008 AVAYA All rights reserved.

SoftConsole Administration

SoftConsole has an administration mode that enables the operator to configure the following settings:

- **Control panel views**
  The BLF panel, call history panel, held calls panel and park slot panel can be hidden or made visible.
- **Change the Administrator password**
- **Edit operator profiles**
  Each operator can have a personalized profile, which can be configured by the administrator.
- **Create and modify templates**
  SoftConsole comes with three predefined templates, which can be modified, or new templates can be created.
- **Specify the maximum length of call notes**
  IP Office supports a wide range of different telephone types. These have different display sizes so the operator can define the character length of messages sent to each user according to the type of phone they use.
- **System Tray working**
  The application can be minimized and left running in the system tray so that it can pop on received calls.

SoftConsole Telephone Requirements

- SoftConsole provides extensive call management, but it still requires an IP Office telephone to provide the speech path. SoftConsole has been tested and is certified to work with all Avaya wired digital and IP phones that are listed in chapter 4.
- SoftConsole cannot be used with IP DECT 3700 series telephones.

SoftConsole PC Requirements

- IP Office software release 2.0 or later.
- PC requirements:
  - Always refer to the latest Avaya SMB Technical Tip or Technical Bulletin for any updated information with regard to Operating Systems, Service Packs or PC hardware
  - Refer to Technical Specifications section of the Product Description for Operating System and Hardware requirements
- A maximum of four SoftConsole applications can be run per system. An IP Office license controls the number of simultaneous SoftConsole users.

© 2008 AVAYA All rights reserved.

© 2008 AVAYA All rights reserved.

Chapter 10. Computer Telephony Integration

© 2008 AVAYA All rights reserved.

10. Computer Telephony Integration

Computer Telephony Integration (CTI) is about bridging the gap between the telephone system and business applications. On IP Office, this is achieved by use of the IP Office CTI Link, a CTI middleware product and Software Developers Kit.

On IP Office, CTI is delivered through adherence to open standards. This gives businesses access to a wide range of third-party solutions, addressing vertical markets, and designed to meet their requirements. For developers, migrating their offering from other platforms to IP Office is quick and easy, and the advanced CTI features IP Office offers makes it easy to demonstrate full integration, and more business benefits.

IP Office provides two levels of CTI interoperability: CTI Link Lite, which is free of charge, provides all the functionality required to support the vast majority of applications, including screen-popping, and many third-party products.

CTI Link Pro provides enhanced functionality, including the ability to control multiple telephones and gives access to advanced call center operation.

Because IP networking is integrated into the IP Office system, all CTI is done through the LAN. On many other systems, CTI is delivered by a physical connection between each handset and computer (first party CTI). This introduces additional points of failure, as well as relying on non-standard interfaces and handsets. On IP Office, all devices can be used with CTI.

© 2008 AVAYA All rights reserved.

CTI with IP Office

IP Office offers a significant CTI capability. Several interfaces are supported:

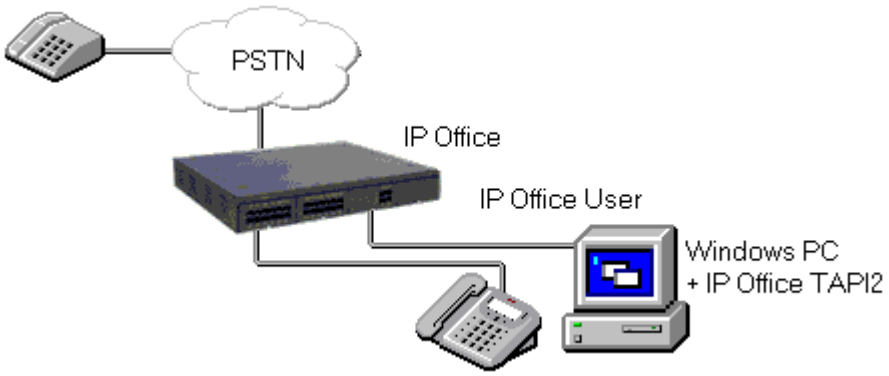
- **TAPILink Lite.**
- **TAPILink Pro.**
- **TAPI-WAV driver.**
- **DevLink Pro.**
- **IP Office SMDR.**
- **IP Office Software Development Kit.**
- **Microsoft™ CRM Integration Phase 1 (Screen Pop).**

- **TAPILink Lite**
  Provides first-party CTI support for Microsoft TAPI 2.1 and TAPI 3.0, so each PC can control or monitor one handset device. The software components are supplied with the IP Office system on the User CD-Rom, and do not required a license key for use.
- **TAPILink Pro**
  Provides third-party CTI support for TAPI 2.1 and 3.0. These components are identical to their first-party equivalent; the presence of the CTI Link Pro RFA license key (which can be purchased in the usual way for products) enables this additional functionality.
- **TAPI-WAV driver**
  Provides software-based support for voice processing. The TAPI-WAV driver is for use with TAPI 2.1 only; for TAPI 3.0, IP Office supports the Media Service Provider (MSP) interface, defined by Microsoft in TAPI 3.0. This functionality will only work in conjunction with CTI Link Pro and also requires TAPI WAV licenses. Each license enables 4 ports of voice processing.
- **DevLink Pro**
  Provides a real-time event stream in addition to the SMDR interface provided in IP Office SMDR. The real-time event stream takes the form of a call record, which is issued whenever the state of any endpoint of a call changes (typically there are two end points on a call, but for some circumstances, such as conference calls, intruded calls there may be more).
- **IP Office SMDR**
  Provides an interface to obtain SMDR events. A comma-separated record is issued for each call, when the call is completed. This interface is designed for call accounting and call billing applications. IP Office SMDR is available free of charge, and distributed on the IP Office Admin CD-ROM.
  - For IP Office 4.2, the IP Office SMDR can be output direct from the IP Office control unit to a designated IP address and port.
- **Software Development Kit**
  This toolkit is delivered on a single CD-Rom, containing the developer documentation for TAPILink Lite, TAPILink Pro, DevLink Lite and DevLink pro, as well as pre-compiled programs for exploring TAPI 2.1 and 3.0. In addition, example source code is included, making it easy for developers to become familiar with IP Office CTI interfaces.

© 2008 AVAYA All rights reserved.

TAPILink Lite (1st Party TAPI Support)

TAPILink Lite provides simple first-party CTI via Microsoft TAPI 2.1 and 3.0. Individual desktop PCs connected to the Local Area Network communicate with IP Office via an IP connection over the LAN. Each PC is capable of controlling one telephone device (see diagram below).



Microsoft TAPI 2.1 and 3.0 are specifications and developers interfaces for controlling and monitoring a telephony device. The specification requires that a certain amount of core functionality is implemented, and additionally defines a series of optional functionality that switch vendors may also implement.

TAPILink Pro (3rd Party TAPI Support)

TAPILink Pro provides all of the features and functionality of TAPILink Lite, but additionally provides third party CTI operation. This means that a single server can control and monitor any number of telephone devices.

In addition, TAPILink Pro provides the ability to monitor and control groups. This allows an application to be notified when a call enters a queue, and can also redirect it to another location.

TAPILink Pro also supports additional TAPI functionality that is not available through TAPILink Lite. This functionality is supported through the LineGetLineDevStatus and LineDevSpecific calls. The additional features are:

- Agent login.
- Agent logout.
- Set and retrieve divert destination.
- Set and retrieve extended divert status (Forward All Calls, Forward on Busy, Forward on No Answer, Do not Disturb).
- Retrieving the extension locale (language).
- Set and clear the message waiting lamp.
- Enable and disable group membership.
- Generate and detect DTMF digits and tones (requires the TAPI-WAV driver).

Support for Developers

The Developer Connection Program ("DevConnect") is the Avaya developer partner program, and is designed for third-party companies who are creating a product for sale, and who wish to receive technical support. Membership of the program is at the sole discretion of Avaya.

DeveloperConnect members pay an annual fee, for which they receive technical support directly from Avaya. In addition, Avaya will perform interoperability testing between IP Office and the member's product, and may also create opportunities for joint marketing, including exhibitions, use of Avaya's logo, and other benefits.

More information on the DeveloperConnect program can be found at www.devconnectprogram.com.

© 2008 AVAYA All rights reserved.

Chapter 11. Voicemail

© 2008 AVAYA All rights reserved.

11. Voicemail

Voicemail provides a telephone answering machine with a personalized greeting on every employee's desk and allows callers to leave spoken messages when the user cannot answer a telephone call. Voicemail messages are retrieved either locally or remotely via any telephone (users are prompted for a PIN if they are using any telephone other than their allocated extension or a trusted location e.g. mobile telephone).

For users that prefer to have email as their main message store, they can forward their voice messages to their email and collect them via their email account.

The voicemail server is multi-lingual and can offer different prompts depending on the user's preferred language, independently of the default system setup. Similarly, external callers can hear prompts in their own language depending on their incoming call route (e.g. based on caller ID).

Four voicemail options are available:

- **VoiceMail Lite**
- **Embedded Voicemail** (IP Office 500)
- **VoiceMail Pro**
- **Centralized INTUITY Audix / Modular Messaging Voicemail**

Positioning Summary

For further details refer to Voicemail Feature Comparison 219 at the end of this section.

Feature	Embedded Voicemail	Voicemail Lite	VoiceMail Pro
Supported IP Office Systems	IP500.	PC Based - All IP Office systems.	
Mailboxes	IP Office automatically creates mailboxes for each user and hunt group on the system.		
Message Storage Capacity	Up to 15 hours.	1MB per minute up to hard disk capacity.	
Maximum Simultaneous Calls (Requires licenses)	4.	4.	IP412 = 30. IP500 = 30.
Centralized operation.	No.	No.	Yes.
Queue Announcements	Yes.	Yes.	Yes.
Auto Attendant	Yes.	No.	Yes.
Call Recording	No.	No.	Yes.
Intuity Emulation	No.	No.	Yes.

© 2008 AVAYA All rights reserved.

Voicemail Lite

VoiceMail Lite is the IP Office basic voicemail application and can handle up to 4 simultaneous calls. Each user has the option of turning their voicemail on or off. When on, the system automatically answers their telephone when they are not available to take a call, plays a personal greeting to confirm that the intended recipient will actually receive the message, and records a message.

When a message has been left, the user will see a message-waiting lamp lit on their telephone and can press a retrieval button to collect their messages.

VoiceMail Lite can ring the extension to deliver any new messages. When voicemail messages are left they are time and date stamped and the caller's number noted. Once listened to, old messages are automatically deleted 36 hours after being left, unless the user chooses to save the message permanently.

Voicemail can be collected remotely by calling into the VoiceMail Lite server. If the number the user is dialing from is recognized (home number or Mobile/Cell Phone for example), they will listen to their voicemail straight away. If the source number is not recognized, users will be prompted for a mailbox number and a PIN code for that mailbox, before they can listen to voicemail. Users have the ability to set and change their own PIN codes.

Where a voicemail needs to be copied to other users, VoiceMail Lite provides many options:

- Voicemails can be forwarded to another mailbox, or group of mailboxes
- Recipients can add their comment to the voicemail before forwarding to another mailbox or mailboxes.
- Voicemails can be forwarded as email WAV attachments.

VoiceMail Lite telephony user interface (TUI) only operates in IP Office mode, not INTUITY mode.

Note: On the IP500, VoiceMail Lite is only supported after upgrading to IP Office Professional Edition.

© 2008 AVAYA All rights reserved.

Embedded Voicemail

(IP500, IP406 V2 and IP Office - Small Office Edition only)

In environments like retail or home office, where space, noise or cost considerations rule out using a PC for voicemail, Embedded Voicemail will be the preferred option for an entry-level voicemail service. With the Small Office Edition Embedded Voicemail makes use of the voice compression resources to optimize the message storage by compressing messages before storing, and expanding them during playback. By doing this up to 10 hours of messages can be stored for all users of the system. Neither the IP500 nor the IP406 V2 require voice compression modules for storing messages and both support up to 15 hours of storage.

Key features of Embedded Voicemail include:

- 3 Port voicemail as standard on Small Office Edition (10 ports with 16VC variants of Small Office Edition), 4 port voicemail for IP500 and IP406 V2.
- Up to 10 hours storage on Small Office Edition, 15 hours message storage on the IP500 and IP406 V2.
- Configurable record time: Default value 2 minutes, maximum value 3 minutes.
- Mailbox security codes ensure a minimum of 4 characters to be set.
- Multiple languages stored on the Flash Memory card.
- Help menus (via *4). Greetings and Mailbox Navigation.
- From Release 4.2, it is possible to send an email containing either notification or the complete voicemail message from the Embedded Voicemail (IP406 V2 and IP500 units only).
- Voicemail Breakout/Personal Auto-Attendant: Up to 3 breakout numbers can be set up. When callers are directed to your mailbox, they can either leave a message or choose to be transferred to one of three numbers (e.g. Operator, mobile/cell phone, colleague, etc).
- Configurable system-wide short code for Voicemail collect (e.g. *17).
- 40 independent Auto Attendants with 3 time profiles per Auto Attendant.
- Up to 12 menu items per Auto Attendant with automatic time-out to fallback number.
- Access and control of voicemail via the digital or IP terminal display (Visual Voice). This feature is supported on the 2410, 2420, 4610, 4620, 4621, 4625, 5410, 5420, 5610, 5620 and 5621 phones.
- Reply to a message to either an internal or external number (if Caller ID available).
- Support for Hunt group announcements.
- Fax option for rerouting fax calls via the auto-attendant menu.
- Support for Fast Forward (#), Rewind (*), Skip message (9) and Call Sender (**) when listening to messages.
- No License Key required.

© 2008 AVAYA All rights reserved.

Voicemail Pro

IP Office VoiceMail Pro offers all the features and facilities of VoiceMail Lite and can be tailored to meet the individual needs of a business. It has higher call capacity by scaling up from 4 to 30 simultaneous calls. All options are available in a choice of languages; both spoken voice prompts and graphical programming interfaces and have the choice of IP Office TUI and INTUITY emulation TUI.

At the heart of VoiceMail Pro is the ability to construct call flows from a series of different building blocks. These building blocks allow automation over tasks like answer a call, listen for tone-dialed digits, make a call etc. VoiceMail Pro call flows allow far more than just guiding a user to the group or extension they require. Call flows allow VoiceMail Pro to dial back users as soon as a voicemail message is left for them, it provides remote access to phone forwarding settings should a user wish to change their Forwarding or Follow Me number from an external telephone. VoiceMail Pro provides message handling for individuals or groups, audio information to callers so assisting the operator during periods of heavy call activity and links to business applications through services such as Text-to-Speech. VoiceMail Pro provides a full telephony applications environment where call flows can be set up and interact in real time with business workflow – callers can interact via menus and data entry and VoiceMail Pro applications can speak back results. For example, users can listen to their email messages through the telephone

A single PC based VoiceMail Pro server can provide voicemail services to multiple IP Office systems in a Small Community Network over the LAN, WAN or a Frame Relay network. This is referred to as 'Centralized Voicemail' and can reduce costs, while facilitating communication between IP Office sites.

Other uses for VoiceMail Pro include:

- Whisper Announce that prompts callers for information (usually their name) which is recorded and passed on to the user's extension on answer, allowing them to choose to accept the call or not. This is particularly useful on "CLI/ANI withheld" numbers - usually calls from telesales companies where somebody is trying to sell you something. VoiceMail Pro will not intrude onto busy extensions.
- Assisted Transfer allows transfer of a call to a destination, but allows the call to return to VoiceMail Pro automatically for other options should the called party be engaged, or not answer within a pre-determined time.
- Conditional routing of calls. Conditions are constructed from a set of basic elements. These elements can be combined within a single condition to create complex rules. For example, the Week Planner can be used to define the company's standard working hours, and then combined with the calendar to define exception days such as public holidays / vacation.
- Call modules. Modules allow you to create sequences of actions that you want to share between a number of different call routing scenarios – like a “macro” in PC applications. These modules can be used to create a library of vertical voicemail applications or just easy dissemination to other IP Office voicemail sites, thanks to its import and export functionality.
- Activation of the external relays on the IP Office system. For example, remotely checking the status of the office heating and then turning it on from your Mobile/Cell Phone on your drive in to work.
- Finally, a Speaking Clock, that takes its time from the voicemail server, is built into VoiceMail Pro to minimize call charges.

© 2008 AVAYA All rights reserved.

Key features of VoiceMail Pro include:

- VoiceMail Pro client, a graphical user interface for programming and configuring applications both locally and remotely.
- IVR for individual business requirements.
- Personal Numbering.
- Broadcast group messages.
- Audiotex and Auto-Attendant services (including dial by name).
- Sophisticated Queue Announcement facilities.
- Conditions (e.g. test if 'out of hours').
- Automatic and On Demand Call Recording with an option for ContactStore Search and replay of saved messages
- Voice Forms/Questionnaire Mailboxes (Campaign Manager).
- Personal distribution lists.
- Access to Database information for building Interactive Voice Response (IVR) systems.
- Tag information retrieved from a database to a call and delivers it with the call to an agent.
- Visual Basic (VB) Script support to allow the configuration of the voicemail system through VB Scripts rather than VoiceMail Pro call flows.
- Extended Personal Greetings to customize the information presented to a caller based upon the availability of a user.
- Text To Speech facilities to allow emails to be read out over the telephone and/or for database information to be read to a caller in 14 languages.
- Housekeeping facilities for the management of messages.
- Automatic detection and routing of Fax calls within Auto Attendants and within a subscriber's voicemail box.
- Forwarding of voicemail messages to Email systems via SMTP.
- Unified Messaging (UMS) offers voice mail - email synchronization between VoiceMail Pro server and email client.
- VoiceMail Pro Web Access allows access to voice mails via a web interface.
- Support for a range of the INTUITY telephone user interface features in INTUITY emulation mode.
- Recording of system prompts through the telephone handset or using multimedia facilities on a PC.
- Speaking Clock.
- 22 supported prompt languages: Chinese (Mandarin), Danish, Dutch, English (UK), English (US), Finnish, French (France), French (Canadian), German, Greek, Hungarian, Japanese, Italian, Korean, Norwegian, Polish, Portuguese (European), Portuguese (Brazilian), Russian, Spanish (Castilian), Spanish (Latin American), Swedish
- Support for TTY hearing impaired text phone
- Centralized voicemail within a multi-site IP Office environment.
- Networked Messaging with other Avaya voicemail systems.
- Capacity of up to 30 ports (depending on IP Office Control Unit).
- Voicemail channels between VoiceMail Pro and the IP Office can be reserved for business critical functions or left unreserved for any function.
- Access and control of voicemail via the digital or IP terminal display (Visual Voice).
- Improved voice recording, including recording of calls made over IP endpoints (including those using Direct Media); automatic call recording triggered by Incoming Call Routes; pausing recording when call is parked or placed on hold.
- User start points in VoiceMail Pro now include Queued and Still Queued options.

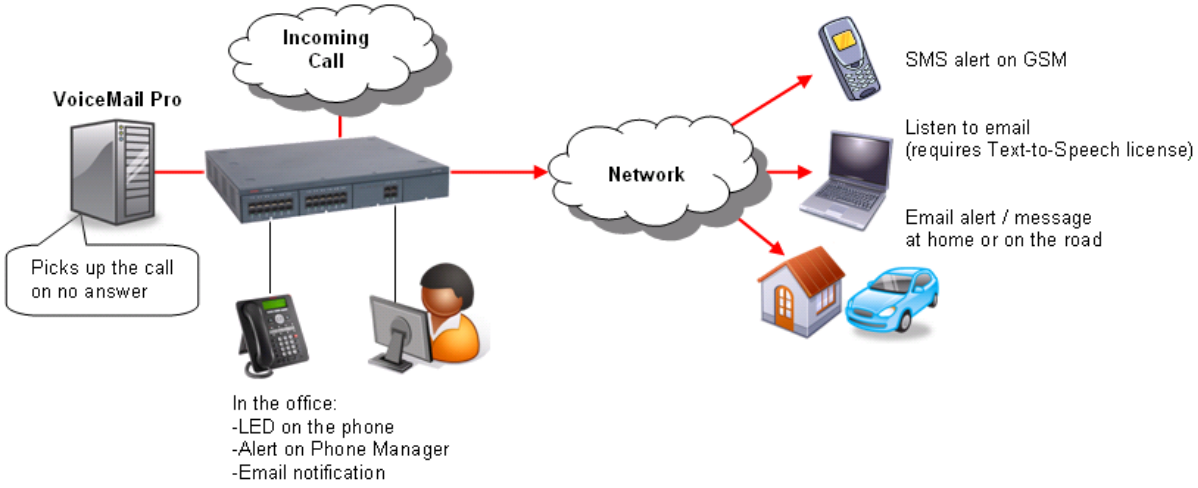
Further details on some of the VoiceMail Pro functionality listed above are described later in this section. Further information on Queue Announcements can be found in Compact Contact Center (CCC).

Note: on the IP500, VoiceMail Pro is only supported after upgrading to IP Office Professional Edition.

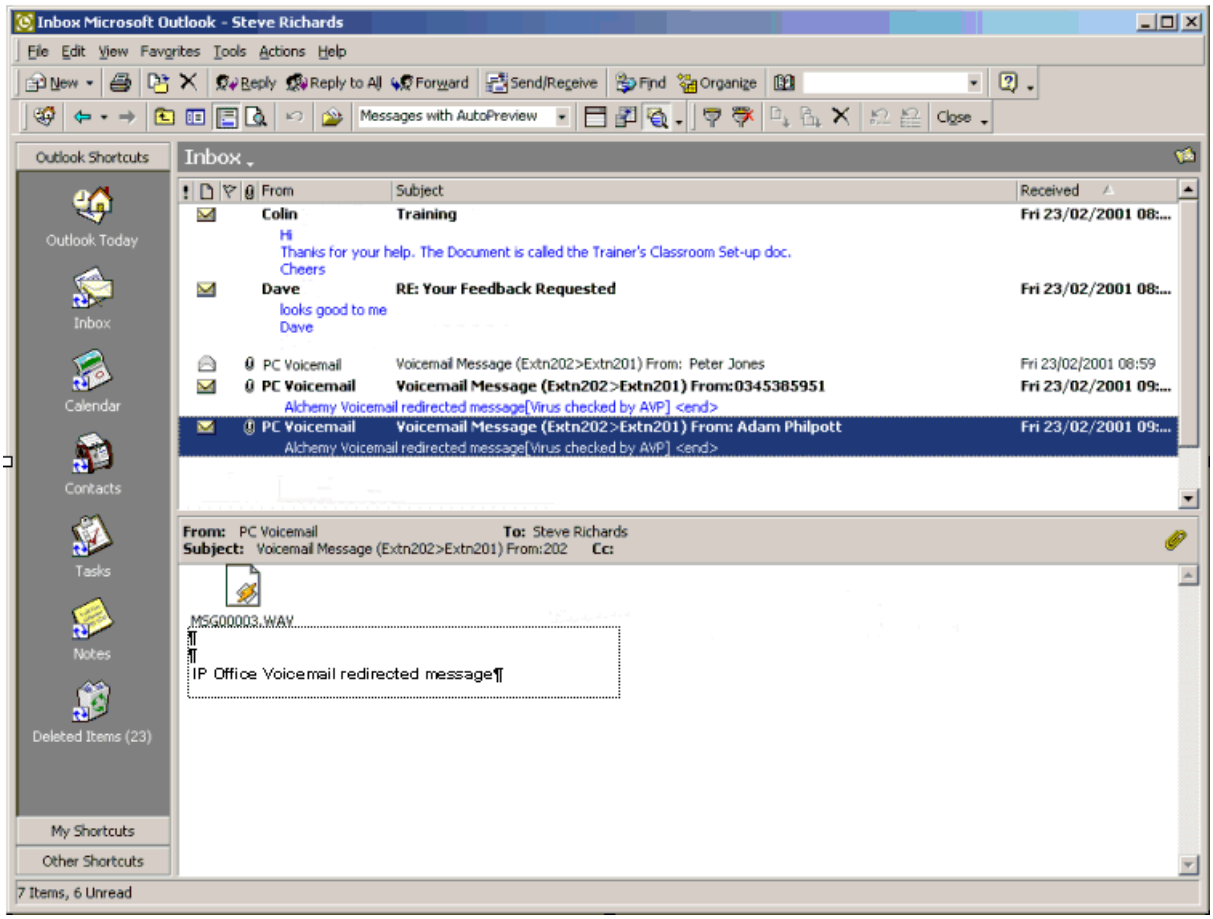
© 2008 AVAYA All rights reserved.

Interaction of Voicemail with Email Systems

As standard, VoiceMail Lite and Pro allow for a simple voicemail alert where the entire voicemail is forwarded (copied) as a .WAV attachment to any MAPI or SMTP compliant Email application. (Microsoft Outlook, Exchange, Lotus Notes, etc.) Forwarding allows emails and voicemail messages to be unified and collected from a single source. This simple alert option that forwards only the caller's number in the subject of the email, and is ideal for use with commercial Short Message System (SMS) or paging services whereby this information can be forwarded to the display on a Mobile/Cell Phone or Pager when the user is away from the desk. This email notification, forwarding and copying, can be done for all voice messages and can be activated remotely. This is beneficial if you are working from home and have an email connection available.


Incoming Call
VoiceMail Pro
Picks up the call on no answer
Network
SMS alert on GSM
Listen to email (requires Text-to-Speech license)
Email alert / message at home or on the road
In the office:
-LED on the phone
-Alert on Phone Manager
-Email notification

Forwarding voicemail to email is one element of unified messaging and is particularly useful for group voicemail boxes as it allows a single voicemail message to be copied to the email of every member in that group.


Inbox Microsoft Outlook - Steve Richards
File Edit View Favorites Tools Actions Help
New Reply Reply to All Forward Send/Receive Find Organize
Messages with AutoPreview Close
Outlook Shortcuts
Outlook Today
Inbox
Calendar
Contacts
Tasks
Notes
Deleted Items (23)
My Shortcuts
Other Shortcuts
Inbox
From Subject Received
Colin Training Fri 23/02/2001 08:...
Hi
Thanks for your help. The Document is called the Trainer's Classroom Set-up doc.
Cheers
Dave RE: Your Feedback Requested Fri 23/02/2001 08:...
looks good to me
Dave
PC Voicemail Voicemail Message (Extn202>Extn201) From: Peter Jones Fri 23/02/2001 08:59
PC Voicemail Voicemail Message (Extn202>Extn201) From:0345385951 Fri 23/02/2001 09:...
Alchemy Voicemail redirected message[Virus checked by AVP] <end>
PC Voicemail Voicemail Message (Extn202>Extn201) From: Adam Philpott Fri 23/02/2001 09:...
Alchemy Voicemail redirected message[Virus checked by AVP] <end>
From: PC Voicemail To: Steve Richards
Subject: Voicemail Message (Extn202>Extn201) From:202 Cc:
MSG00003.WAV
IP Office Voicemail redirected message
7 Items, 6 Unread

Presentation of Voicemail to Email

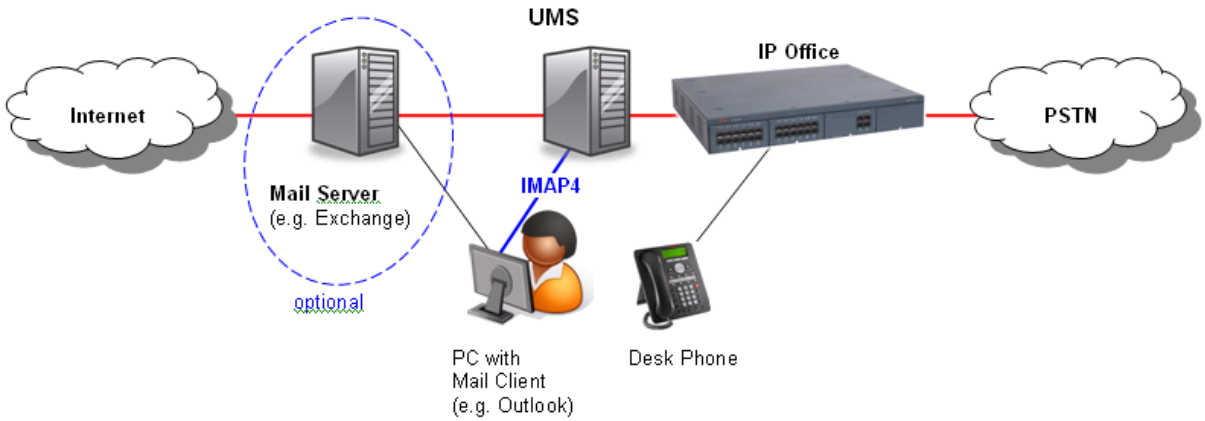
© 2008 AVAYA All rights reserved.

Unified Messaging Service (UMS)

Unified Messaging Service (UMS) is being introduced to replace the Integrated Messaging Solution (IMS). While it delivers almost the same functionality that was available on IMS, the UMS solution dramatically simplifies the installation and configuration. It delivers the user with two client interfaces through which voicemail messages can be retrieved from VoiceMail Pro.

Voice Mail Synchronization via IMAP

Email applications such as Outlook which support IMAP can connect to an IMAP server integrated with the VoiceMail Pro server.



When using a mail client new voice mails arrive as an email with wave attachment and the user can handle them like every other email. If a voicemail has been listened to, its status will change to "read" everywhere. If the user deletes it, it will be deleted everywhere.

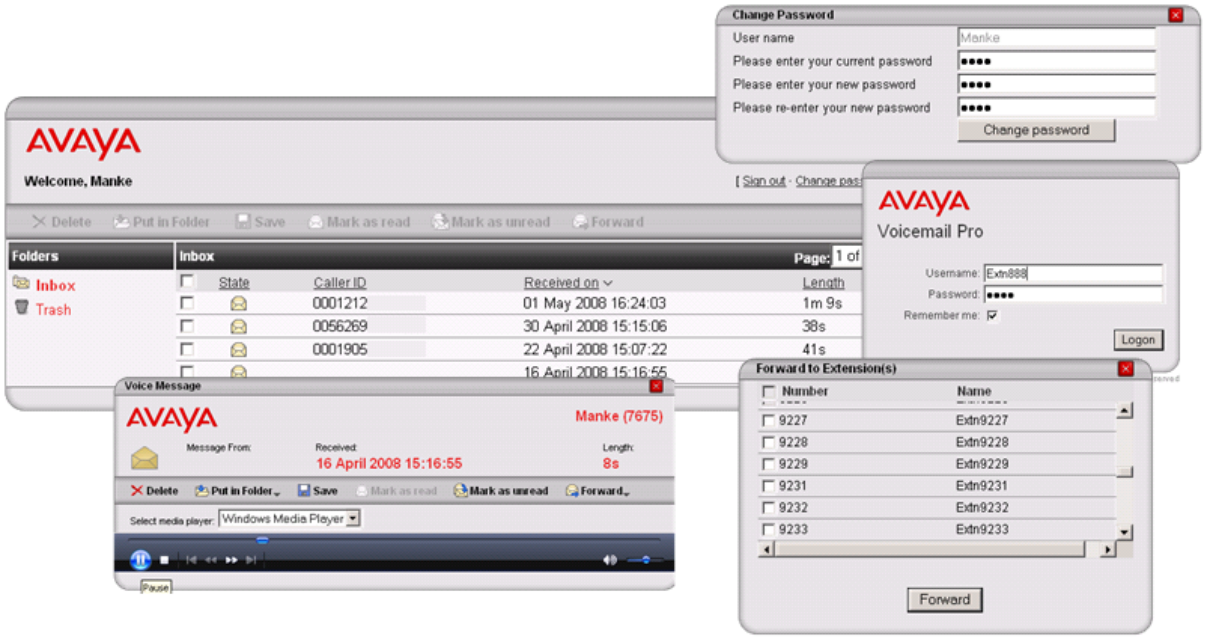
Users that have UMS access enabled just need to add an additional email account to their client to be able to use the unified messaging features.

The instructions how to add this account and configure the web address for the VoiceMail Pro Web Access can be provided by an administrator email without the need for system administration or Business Partner support.

© 2008 AVAYA All rights reserved.

VoiceMail Pro Web Access

The second interface the UMS license offers to the users is a web interface allowing voicemail access via a web browser, such as IE7 or Mozilla Firefox, with the ability to listen to targeted voicemails by using either the PC's multimedia equipment or the desk phone.



Unified Messaging Services web services requires the VoiceMail Pro installation to be on a server with IIS pre-installed. In addition, the VoiceMail Pro web services option must be selected during installation.

Once installed, the system will require the following additional configuration:

- UMS licenses (for the number of users that will be configured to use UMS)
- the users selected for UMS support must be enabled by the system administrator in Manager
- each user must have a voicemail passcode (a blank passcode is not acceptable)

As the UMS solution is not reliant on integration with Exchange significant savings are made in the cost of installation and maintenance.

© 2008 AVAYA All rights reserved.

Fax Messages

While not directly supplying or supporting fax software, integration with fax to the desktop or client fax applications can be done through the use of fax servers. This then allows an email client (for example Microsoft Outlook) to be utilized as an easily affordable unified messaging solution. The many benefits of unified messaging include security (as faxes are sent to the users PC rather than on paper for everyone to see), ease-of-use and efficiency in terms of storage and retrieval of messages and the great gains that can be made in overall workforce efficiency and productivity.

To enhance the support of third party fax solutions, VoiceMail Pro supports the automatic detection of incoming fax calls. Traditionally a dedicated telephone number is provided for all incoming fax calls. In addition to, or as an alternative to, the VoiceMail Pro 'Menu' action or a subscriber's voicemail box (INTUITY mode) can automatically detect any incoming fax calls and then direct the call to a predefined location. The benefit to a business or user is that only one number is required for either voice or fax calls.

The VoiceMail Pro can store the default fax location for the automatic routing of fax calls. Alternatively, with fax tone detection at the voicemail box, each voicemail box can have a fax location number. If a voicemail box owner has set his or her own fax number, then that number is used instead of the default fax location.

Voicemail box subscribers can set their own fax number through their mailbox menus.

Most fax solutions can be used in conjunction with IP Office, however the following products have been tested and verified to operate in the above scenarios:

- **Equisys - Zetafax**
  Zetafax for Networks provides versatile network fax software solutions for small businesses, corporate offices and distributed enterprise businesses. It enables employees to send and receive faxes at their desktop, without the need to print fax communications, take them to a fax machine and send them manually. Zetafax can be seamlessly integrated into market leading email systems like Exchange allowing users to send and receive faxes directly from their Outlook client. In addition Zetafax can be integrated with other existing applications, such as accounting or CRM systems, for fast, automated faxing from the desktop or back office. Zetafax for networks is already used by more than 60,000 customers worldwide. Further product information available from www.equisys.com
- **Captaris - RightFax**
  RightFax offers a broad, scalable product line that integrates with email, desktop, CRM, ERP, document management, imaging, archival, call center, copier/scanner systems, as well as host, legacy and mainframe applications–virtually all business applications. Further product information available from www.captaris.com
- **Fenestrae – Faxination**
  Fenestrae Faxination Server for Microsoft Exchange integrates fax into email technology. Create faxes on your desktop and deliver them to your chosen fax machine at the click of a mouse. Further product information available from www.fenestrae.com
- **GFI – GFI FaxMaker**
  GFI FaxMaker for Exchange/SMTP allows users to send and receive faxes and SMS/text messages directly from their email client. It integrates with Active Directory and therefore does not require the administration of a separate fax user database. GFI FAXmaker integrates via the SMTP/POP3 protocol with Lotus Notes and any SMTP/POP3 server. Further product information available from www.gfi.com
- **Castelle Fax**
  Faxes routed to a user's mailbox by Castelle fax servers will be recognized by VoiceMail Pro as faxes, and will be supported by VoiceMail Pro Fax features.

© 2008 AVAYA All rights reserved.

Email Reading (Microsoft Exchange only)

In addition to providing a unified mailbox for voicemail messages, emails and fax message, VoiceMail Pro can also provide the ability to retrieve email messages through the telephone. When operating in INTUITY mode and with the system licensed for Text-To-Speech (TTS) facilities the user will be presented with a list of both voicemail messages and email messages. The emails can be read out over the telephone in any of the supported 14 languages, based upon the system or user localization settings. The benefit to the user is that their messages are now accessible while in and out of the office through any telephone.

When accessing messages through the telephone all new voicemail messages will be presented to the mailbox owner before any new email messages. When accessing an email message the system refers to the message as *"New message with text"*.

Configuring the reading of emails to users is a simple exercise. Firstly, TTS services will be loaded onto the VoiceMail Pro server (the Avaya TTS media pack will install the Avaya TTS engine). Secondly, a TTS license key will need to be purchased and entered into IP Office manager. Thirdly, for each user who is wishes to utilize email reading, the user's email address will need to be entered into the user profile details in IP Office Manager and the facility enabled through the email reading checkbox.

Where the user has email reading in their voicemail box, they will be able to record a voice reply to the email, and send it as a .WAV attachment to a reply email to the person who sent the email.

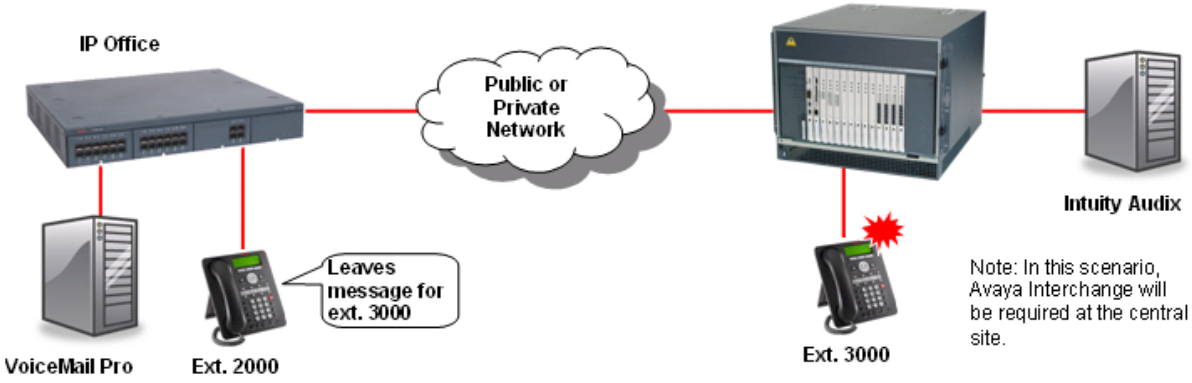
© 2008 AVAYA All rights reserved.

Networked Messaging

Where organizations are operating a number of voicemail systems across different sites it is important to be able to provide integrated operation between voicemail systems so that messages can be passed between systems and delivered to a user's mailbox seamlessly. This is achieved by IP Office VoiceMail Pro being licensed to support Networked Messaging.

The Networked Messaging Solution defines a common set of features to allow inter-working between Avaya voicemail systems. In INTUITY mode, while listening to or having listened to a message, the user can select the option to forward the message to another mailbox, the mailbox entered can be any mailbox number on the local system or any mailbox on a remote Avaya system.

The IP Office Networked Messaging facility will allow configuration of up to 2000 remote mailboxes on each VoiceMail Pro server and will operate with other IP Office systems supporting this feature, as well as the Avaya Interchange and Avaya S3210 servers.


IP Office
Public or Private Network
Intuity Audix
Leaves message for ext. 3000
VoiceMail Pro
Ext. 2000
Ext. 3000
Note: In this scenario, Avaya Interchange will be required at the central site.

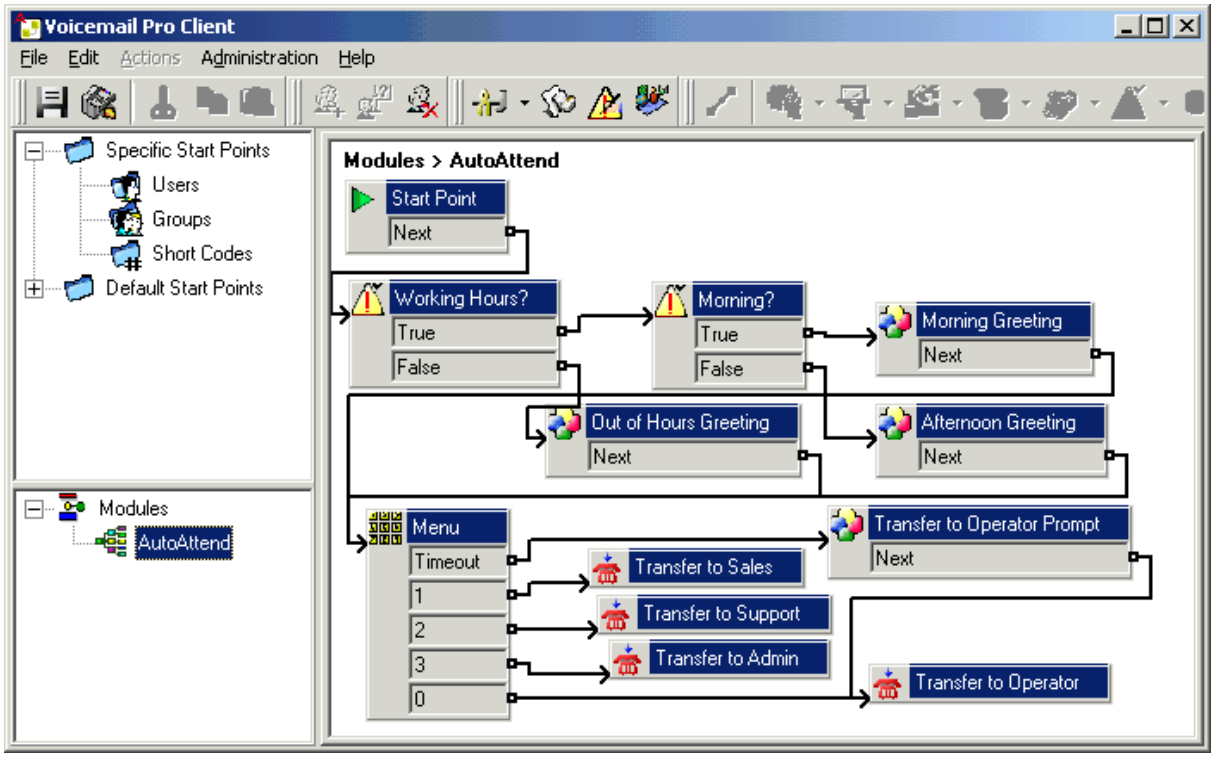
© 2008 AVAYA All rights reserved.

Auto Attendant

VoiceMail Pro provides an easy-to-use, multi-level configuration tool (the VoiceMail Pro client) which allows network managers and system administrators to construct an interactive menu system, based upon DTMF telephone key entry. This allows an Auto-Attendant system to be built and configured to suit business needs, be that on its own or as a back-up for the regular operator when call volumes are high. VoiceMail Pro offers the caller the ability to dial the name of a person via the phone keypad (like "Text" messaging on cell/mobile phones). In response the auto-attendant offers the caller a best match name or if there is more than one, a selection list is offered and the caller can select which one they want to call.

As an example, VoiceMail Pro can be used to build an Auto-Attendant that prompts callers to "enter 1 for sales, 2 for support, 3 for admin, or 0 for the operator" allowing them to be transferred to the appropriate department without operator intervention. Alternatively, a list of personnel and their extension numbers could be listed, allowing the caller to directly access the person they want. For larger companies it could be department name listed first, followed by the list of employee extensions within the department.

The latter two examples are ideal where company telephone operation has changed from a central operator to Direct Dialing (DDI/DID), allowing callers to "learn" the required extension number from the prompting of VoiceMail Pro, and then in future dial the extension number, or other pre-defined variables, directly. Auto-Attendant operation is also ideal where multiple languages are required, for example "Dial 1 for English, 2 for German, 3 for French, ...".



Auto-Attendant created using VoiceMail Pro Manager

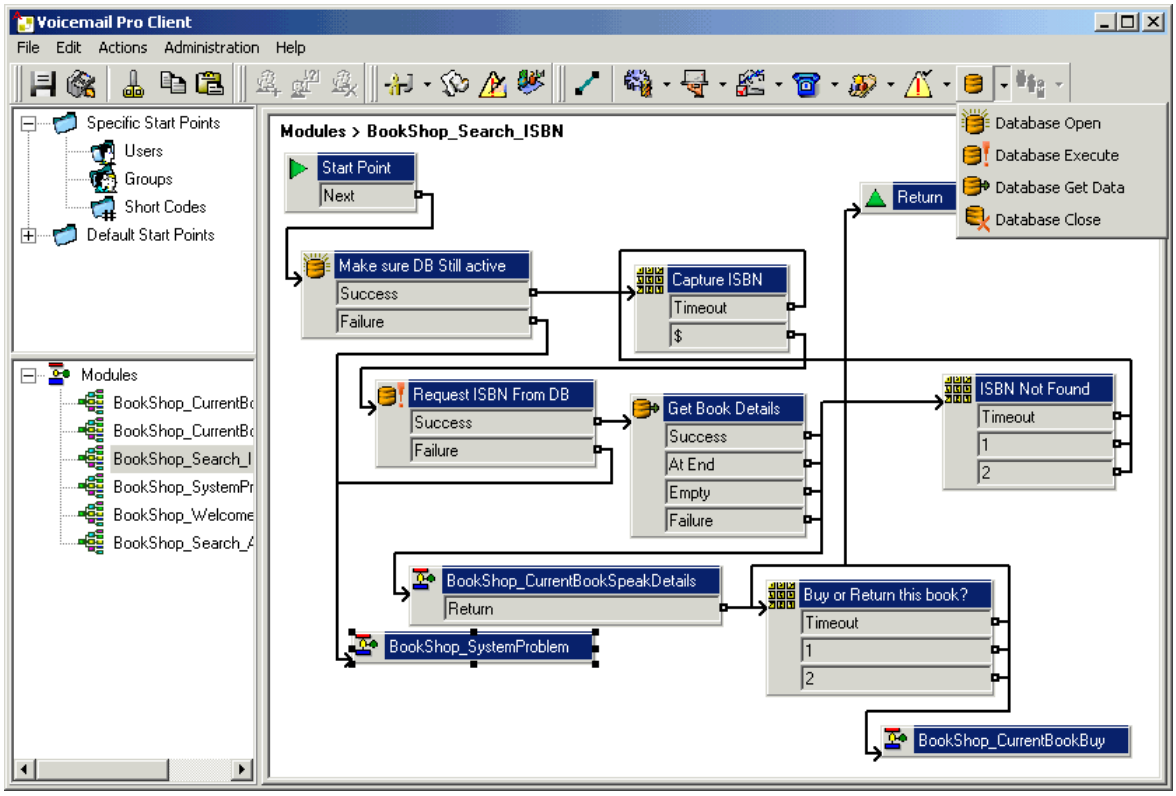
© 2008 AVAYA All rights reserved.

Accessing Database Information within Call Flows (IVR)

VoiceMail Pro provides the ability to construct powerful interactive systems based upon DTMF telephone key entry. This is achieved by using the flexibility provided from the built-in call flow actions. As a caller passes through any part of a defined call flow the system is capable of interacting with most third party databases using the standards based ADO interface (ActiveX Data Objects). The system is capable of retrieving information from a database and writing information into databases. The result of this is that powerful Interactive Voice Response systems (IVR) can be delivered to specifically meet the requirements of the business and the customer experience that is required.

Example interactive systems that can be built as a result of these facilities include: Information Bulletin Boards, order taking and order processing systems, front end systems to Help Desks/Support Desks, Contact Centers, secure access to information through PIN checking, survey systems, remote time sheet management, etc.

- The ability to interact with Database information is enabled through the purchase of the IPO LIC - IP400 3rd PRTY IVR RFA license key. The entry of this key will enable the operation of four new Database Action Icons within the VoiceMail Pro client.



Example Call Flow Utilizing Database Actions

The database actions that are provided through the VoiceMail Pro Client are:

- Database Open – Opens a link to the required database. Multiple databases can be accessed during a call but only one database can be opened at one time.
- Database Execute – Provides the ability to enter a query on the opened database. The query can 'Select' data from the open database or can 'Insert' data into the database.
- Database Get Data – Provides access to the data that has been retrieved from a database through the Database Execute action. The user can retrieve the next item, previous item, first item in the list or the last item in the list.
- Database Close – This action will close the current database. If the database is open when a call terminates then the database will be automatically closed.

As with other VoiceMail Pro call flow actions, the database actions include the ability to communicate with the Avaya Compact Contact Center for reporting purposes, the VoiceMail Pro installation includes Microsoft Data Access Components (MDAC) to simplify connection to most common databases.

Interaction with the opened database is done through Structured Query Language scripts (SQL). An administrator can enter SQL script directly into the specific section of the Database Execute action. For administrators that are not familiar with SQL scripts, a script can be created automatically through the use of a SQL Query Builder Wizard.

© 2008 AVAYA All rights reserved.

Using Text To Speech (TTS) Facilities within a Call Flow

A Text To Speech (TTS) engine can be added to further enhance IP Office IVR capabilities; TTS facilities can enhance the callers experience by allowing the system to read back to them any information that has been extracted from a database. For example, in a Book Shop, the caller dials into the system and is asked for an ISBN number of the book they require. The caller enters the ISBN through the telephone keypad and the system locates the title of the book from the database. As well as finding the title, the system could also look up the author of the book and whether there were any books in stock. By using TTS, the system could now respond to the call:

"The book, Lord Of The Rings, costing $6.99, written by J R R Tolkien is in stock".

The languages currently supported by the Avaya TTS engine are:

- Chinese (Mandarin)
- Dutch
- English (UK)
- English (US)
- French (Standard)
- German
- Japanese
- Italian
- Korean
- Norwegian
- Portuguese (Brazilian)
- Russian
- Spanish
- Spanish (Latin)

TTS Licensing

TTS is an optional licensed component of VoiceMail Pro, and adds a TTS resource pool for VoiceMail Pro to use and release as required. TTS licenses are independent of VoiceMail Pro licenses. If a system integrator wants to use a different TTS language set from those supplied by Avaya this can be done by using the 3rd party TTS license instead of the Avaya language TTS. Both license types are based on a concurrent usage model

Visual Basic (VB) Scripting

The VoiceMail Pro call flow programming interface has been extended to allow an administrator to provide Visual Basic (VB) scripted logic that can be interpreted by the VoiceMail Pro server. This ability allows system administrators to program the voice system via VB Scripts thus providing additional choice and flexibility in providing IVR applications. The VB script action contains a VB-Scripting parser (Syntax checker) to ensure the legitimacy of the administrator derived VB Script before it's incorporation. Each VB script action used within a call flow can contain a maximum of 10,000 characters; however a call flow may contain multiple VB script actions within it.

VB Scripting on IP Office VoiceMail Pro is an optional licensed component.

© 2008 AVAYA All rights reserved.

Personal Numbering

Contact-ability is all-important in winning and maintaining business. VoiceMail Pro offers users the ability to remotely turn their voicemail on or off, set their voicemail email forwarding, edit their call forwarding and follow me numbers. Together these actions provide a comprehensive Personal Numbering service for the user who needs to remain in contact regardless of their physical location.

Users with Mobile Twinning are able to remotely activate their twinning capabilities through VoiceMail Pro call flow.

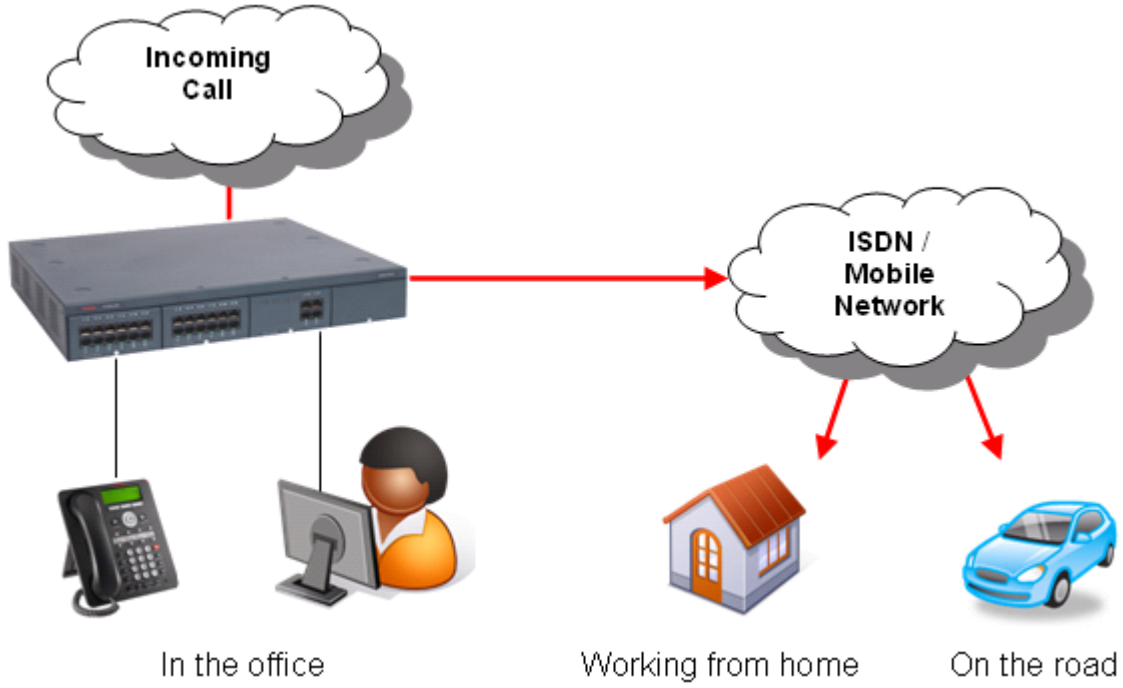


Diagram illustrating personal numbering

© 2008 AVAYA All rights reserved.

Extended Personal Greetings

In INTUITY emulation mode, the VoiceMail Pro system has the ability to hold a number of greetings within each user's mailbox that can be played to a caller. In addition to the standard mailbox greetings, the extended personal greetings provide the ability to present the caller with a greeting that reflects where the call has come from (internal or external) or why the called party is unable to take the call. A mailbox user can configure the responses played back to the caller, based upon the reason the caller was routed to the voicemail. The supported call states are:

- **Busy/Engaged**
  The user is currently on a call and unable to accept a second call.
- **No Reply**
  The user is away from the desk and unable to take a call.
- **Internal**
  A greeting to be played to internal calls
- **External**
  The greeting to be played to external callers
- **Out Of Hours**
  The greeting played when a hunt group is operating 'out of hours'. Out of hours is defined with IP Office Manager and is only applicable to Hunt Group mailboxes.

A greeting can be recorded for each of the above conditions through the Telephone User Interface (TUI). If a recording is made for each condition, the order of play back to a caller will be:

1. Out of hours (Hunt group mailboxes only).
2. Internal/External greeting.
3. Busy/Engaged.
4. No reply.

A mailbox owner will need to record greetings against these conditions to deliver the greeting that they wish to present to a caller. Phone Manager Pro users can record and manage their voicemail greetings through the Phone Manager GUI

© 2008 AVAYA All rights reserved.

Hunt Group Broadcast Messages

With VoiceMail Pro, two modes of operation exist for the handling of hunt group messages. The method used is configured for the group through the IP Office Manager.

- **Hunt group mode**
  Messages are stored in the Hunt Group mailbox and Message Waiting only informs those individuals configured for message waiting indication from that group. This is ideal for scenarios where only a few people such as a call center supervisor need to be initially aware of group messages. Any message waiting light lit by this is extinguished when the new hunt group message is accessed by a user. This is the default mode of operation.
- **Broadcast mode**
  Messages are not stored in the hunt group mailbox. Instead they are broadcast (copied and forwarded) to the individual mailboxes of the entire hunt group membership. This lights the individual messages waiting light of each user of the Hunt Group until they access their mailbox.

Personal Distribution Lists

Personal Distribution Lists are only available with VoiceMail Pro when operating in INTUITY emulation mode. The feature provides the ability for a user to distribute a voicemail message to a list of recipients simultaneously. Lists can be configured by a voicemail box subscriber either through their voicemail box telephone user interface (TUI) or through the desktop PC application Phone Manager Pro.

The features available to a voicemail box subscriber include:

- Create up to 20 lists with 360 members per list
- Mark a list as Private or Public, Private lists can not be accessed by any other voicemail subscriber. Public lists can be used by other subscribers but can not be edited.
- Public lists can be copied from one subscriber to another by adding the contents into a new list.
- Subscribers can 'Create' new lists, 'Scan' contents of an existing list or 'Modify' existing lists.
- List members can be added by using the station number or mailbox name (names are not supported for VoiceMail Pro Networked Messaging mailboxes).
- Lists can include voicemail boxes that exist on other Avaya voicemail systems that are available through VoiceMail Pro Networked Messaging.
- Lists can be added together, duplicate members are automatically removed. This includes public lists owned by other voicemail subscribers.
- Mailing lists are accessible to the user at any 'send message' and 'forward message' option within the user's voicemail box.
- When displayed within Phone Manager Pro, distribution lists can have a list description added to it, this is only visible within Phone Manager Pro.

© 2008 AVAYA All rights reserved.

Cascaded Out-Calling

VoiceMail Pro can send a notification, with an escalation capability, that a new voice message has been received in a user's mailbox to specified phone number(s). This is particularly useful in environments such as healthcare and support where important voice messages are left and need to be answered promptly - even outside of office hours.

For example should a patient leave an important message to the main number of the doctor’s office, the voicemail system can ring the doctor at the office then on no response escalates to the doctor’s mobile/cell phone, his/her home phone or the doctor on duty after a programmable timeout. This avoids having to rely on an external answering service and allows mobile/cell and home phone numbers to remain private.

The voicemail notification can be sent for:

- Any new voice messages
- Any new priority voice messages

Mailbox owners can configure their own options from their handset (Telephone User Interface or TUI) and through their IP Office Phone Manager.

- Create own Time Profile – defining when notification should take place (e.g. office hours only)
- Out-calling destinations – defining where notification should take place and in which priority order

Five destinations can be defined by the mailbox owner through the TUI (Telephone User Interface). The destinations selected in the escalation list are called in sequence. The possible destinations are:

- Desk
- Mobile/Cell
- Home
- Delegate
- Other

Each time an outcall event occurs, each number in the escalation list will be called until either the call is answered, or the end of the list is reached. This process will be repeated on each retry attempt, for the number of retries set.

Out-calling preferences are set for global operation via the VoiceMail Pro Client. Out-calling is only available in INTUITY Mode. The administrator sets the number of retries and time interval between retries on a system-wide level.

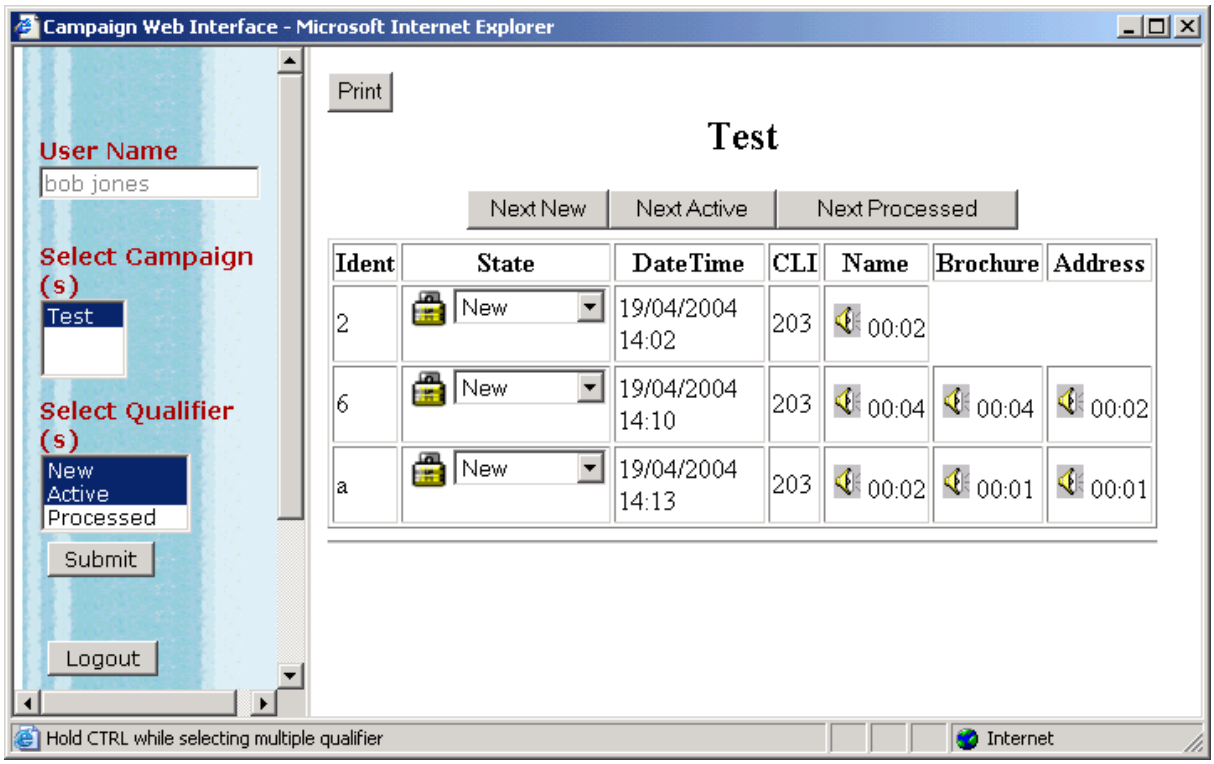
© 2008 AVAYA All rights reserved.

Campaign Manager

As part of VoiceMail Pro, Campaign Manager enables the gathering of repetitive information form inbound calls (such as brochure requests) to be fully automated, leaving agents free to deal with other more complex calls which require human interaction. A definable sequence of recordings are played to the caller with time in between each recording to allow the capture of the caller's spoken answers and/or the caller's key presses via DTMF. At the end of the transaction the caller can be thanked and the completed transaction retrieved by an agent via a web interface or a short code.

Campaign Manager allows calls in queue to break out of the queue, or be directed in an overflow situation to complete their transactions thereby increasing customer satisfaction by effecting an answer to their call. This ensures that a minimum of customers give up when forced to wait in a queue or even worse, hear a recorded message stating that they are calling outside of office hours.

In a Contact Center environment, when agents are busy, an overflow to Campaign Manager relieves congestion and pressure on agent groups. An agent can collect the completed transaction via a web browser or via a short code representing the park slot number of a particular campaign. This number can be pre-programmed under a DSS key and used by agents to access the campaign. If the DSS key incorporates a BLF lamp, that lamp is lit when new campaign messages have been left. Agents then transcribe the caller's answers into a database or other records.



© 2008 AVAYA All rights reserved.

Call Recording

VoiceMail Pro also offers call recording services that allow the automatic/manual recording of calls for a variety of applications, such as for training purposes or to monitor abusive callers. As standard, recordings can be directed to the called extension's voicemail box or to any other mailbox for later retrieval. Alternatively, recordings can be stored in a central database for retrieval through a Web based browser by using ContactStore for IP Office.

The system administrator can select whether all calls are required to be automatically recorded or just a selection of calls. Alternatively, calls can be manually selected for recording. If for any reasons resources are not available then a recording may not be taken (for example all voicemail ports are busy).

VoiceMail Pro provides a number of methods for triggering the recording of a call.

Most of the settings and controls for automatic voice recording are accessed through the IP Office Manager application. The proportion of incoming and/or outgoing calls that should be recorded and the time-period during which Voice Recording should operate can be selected.

- **User Recording**
  The calls to and/or from a particular user can be automatically recorded. By default the recordings are placed in the user's mailbox
- **Hunt Group Recording**
  The calls to a particular hunt group can be automatically recorded. By default the recordings are placed in the hunt group's mailbox, but there is the ability to select a target mailbox made for or on behalf of a subscriber.
- **Account Code Recording**
  An account code can be applied to a call by the user before it is made. This can be used to trigger recording of outgoing calls.
- **Caller ID Recording**
  Account codes can be assigned to a call by Caller ID matching. This allows recording to be based on a Caller ID match.
- **Time Profiles**
  For each user, hunt group and/or account code, an IP Office time profile can be used to determine when auto-recording is used.
- **Incoming Call Routes**
  Incoming Call Routes can trigger automatic call recording.

Note: It is possible for several recordings to be made of the same call. For example, if both automatic hunt group recording and automatic user recording are applicable to the same call, separate recordings are produced for both the hunt group and the user. Recording only continues while the party triggering the recording is part of the call, for example:

- Recording triggered by a user stops when that call is transferred to another user.
- Recording triggered by a hunt group continues if the call is transferred to another member of the same group.
- Recordings triggered by an incoming call route last until the call is cleared from the system.

Call recording uses the conference facility and so is subject to the conference restrictions of the IP Office system. For some situations, it may be a requirement that call parties are advised that their call is about to be recorded. This is done by switching on the Play Advice on Call Recording option via the VoiceMail Pro client. The maximum length of any call recording is 60 minutes

© 2008 AVAYA All rights reserved.

IP Office ContactStore

The standard Call Recording facilities provided with IP Office and VoiceMail Pro can be extended further by using IP Office ContactStore. IP Office ContactStore stores and catalogs recordings so that they are easily accessible for later retrieval. Any recordings that you instruct VoiceMail Pro to "send to the Voice Recording Library" are placed in a database.

IP Office ContactStore is provided with the VoiceMail Pro software CD set and has an inbuilt 45 day trial license. A fully featured IP Office ContactStore system can be installed and used for 45 days from the creation of the first recording. After this time the system will stop taking recordings until a license is purchased and installed onto the IP Office.

IP Office ContactStore has a number of components, these are:

- An MSDE database into which details of all recorded calls are inserted.
- A browser-based call search and replay application.
- A browser-based system configuration and status monitoring application.
- Disk space management - Oldest recordings are automatically deleted as needed.
- Optional archive management - Recordings are automatically written to a DVD +RW drive.

To allow you to search for calls easily, the details of the recordings are stored within a MSDE database. It contains one record for each call recorded and additional records for each party on the call and the owner of the call. The information that is held for any recording is:

- A unique reference for the recording
- The start date and time
- The duration of the recording
- The name and number of the parties on the call—where this was available to IP Office (through ANI, Caller ID or DNIS) at the time of the call.
- The direction of the call (incoming, outgoing, or internal)
- The owner of the call recording
- The target or dialed number, which may be different from the number that actually took the call.

Recordings within IP Office ContactStore are stored as .WAV files. IP Office ContactStore uses the G.726 16kbps ADPCM compression standard, which provides the best compromise between storage capacity and CPU loading. IP Office ContactStore is designed to perform compression as a background task, which does not impact the systems ability to record, search or play other calls. It takes approximately 1 minute to compress a two hour recording. The compressed recordings are stored as 16kbps G.726 format, storage requirements are therefore 8MBs per hour of recording.

The IP Office ContactStore suite can be installed onto the same server as VoiceMail Pro but must be loaded onto a separate partition. Alternatively, IP Office ContactStore can be installed on a separate drive within the same server or on a separate server. The minimum PC specification when VoiceMail Pro and IP Office ContactStore are installed on the same server is detailed in the Voicemail System requirements later in this chapter.

IP Office ContactStore stores recorded calls with certain security in place. Access to recordings is strictly controlled according to the security constraints configured within the System Administration pages. Each recording has an owner; the call owner is the number of the extension that recorded the call. You can specify to which extensions each user has replay rights; the user can search for and replay all calls "owned" by those stations. Typically an individual may be given rights to replay calls owned by their extension number while managers may have rights to the extension numbers of all of their staff.

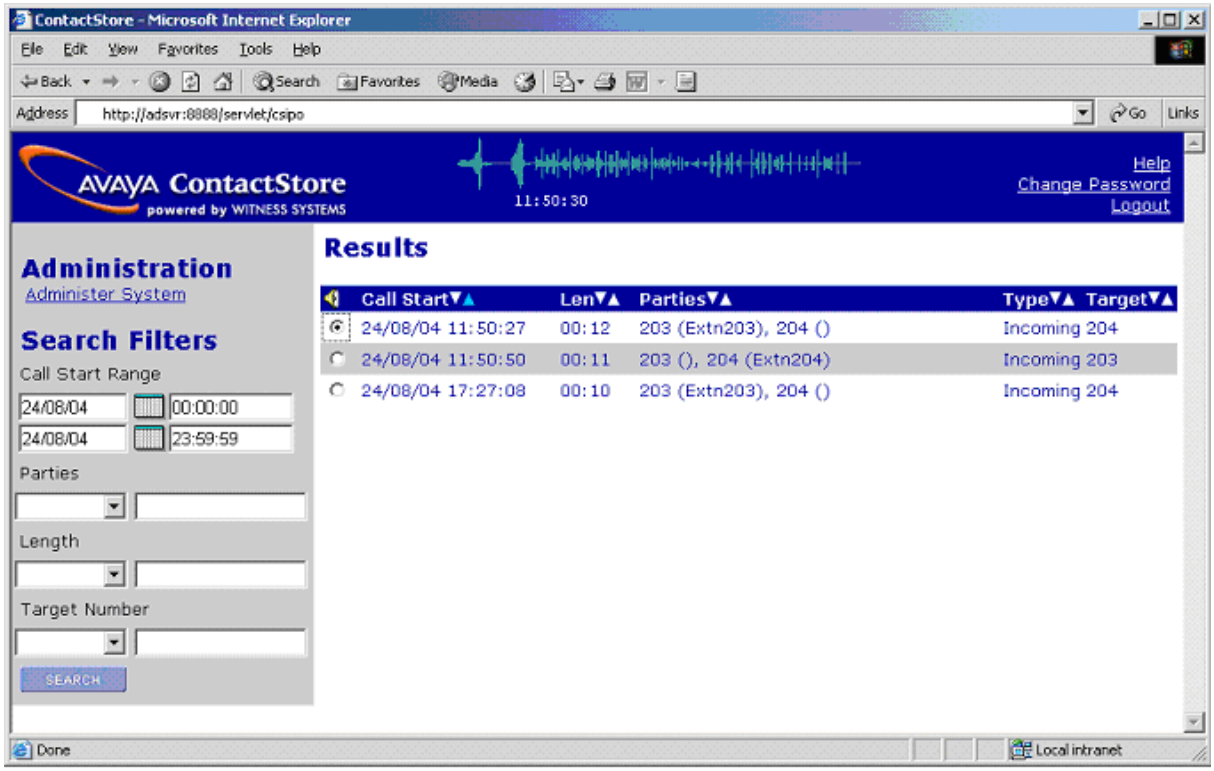
The system will automatically generate alarms showing system warnings. Alarms are logged to IP Office ContactStore's database and held for a month before being purged. The administrator can define specific Email addresses for alarms to be automatically forwarded to. The email recipient could be a local system administrator, a manned help-desk and/or suppliers' support desks if you have a support agreement that includes this facility. The system sends an email message each time an alarm occurs or is cleared. It also sends an email once per day as a "heartbeat" to let you know it is still operating. Failure to receive the daily heartbeat message should be investigated; it could indicate that the server has failed.

© 2008 AVAYA All rights reserved.

IP Office ContactStore allows replay of recordings by means of a browser-based application that is accessible with Internet Explorer (IE) V5.0 and higher. The Search and Replay facilities include the following features:

- Personal security restrictions. The restrictions are applied as you log into the web server.
- Criteria-based search filter fields to perform specific searches.
- Replay controls. Use the replay controls to start, stop, pause, skip forward, skip backward, or to export the recording to a readily playable .wav file.
- Audio waveform display. The waveform presents a graphic representation of the audio content of the call. Use the waveform to avoid replaying static or silences, and to move easily to specific portions of a call.

The Search and Replay screen, shown below, provides filter fields that you can use to search for calls:



Centralized Messaging with Avaya Communication Manager

Where IP Office is deployed in an Avaya Communication Manager (ACM) Environment, it may be desirable to use one centrally managed voicemail system (INTUITY or Modular Messaging) to provide voicemail services to IP Office users. IP Office can be configured to use an INTUITY or Modular Messaging system over a remote connection such that all messaging calls divert to this location and message waiting indications are provided from the remote location and are displayed correctly on IP Office extensions. Connectivity must be either an E1 or T1 circuit or an IP trunk running QSIG services. In addition to the IP Office license Key (Centralized VM with ACM RFA) that enables this service, further license keys may be required on the ACM system.

© 2008 AVAYA All rights reserved.

Voicemail Feature Comparison

Platform Support

	Embedded Voicemail	VoiceMail Lite	VoiceMail Pro
IP500	Yes (does not use VCM resources)	Yes*	Yes*
IP412	No	Yes	Yes
IP406 V2	Yes (does not use VCM resources)	Yes	Yes
Small Office Edition	Yes (uses in built VCM resources)	Yes	Yes

*IP500 running in IP Office Professional Edition mode only.

Capacities

Voicemail	Embedded Voicemail	VoiceMail Lite	VoiceMail Pro
Number of Mailboxes supported	No specific limit on IP500, IP406 V2 or Small Office Edition. Limited only by IP Office configuration.	No Limit - Limited only by IP Office configuration.	No Limit - Limited only by IP Office configuration.
Maximum Number of Concurrent Calls (ports)	4 simultaneous calls on IP500 and IP406 V2. From 1-10 simultaneous calls on IP Office - Small Office Edition depending up on available VCM resources	4 simultaneous calls on IP Office - Small Office Edition, IP406 V2 and IP412	Up to 30 dependent on license and platform (IP Office - Small Office Edition=10, IP406 V2 =20, IP412=30, IP500 = 30).
Recording Time	IP Office 50 and IP406 V2: 15 hours. IP Office - Small Office Edition: 10 hours.	PC dependent (Requires 1MB per minute)	PC dependent (Requires 1MB per minute)

© 2008 AVAYA All rights reserved.

Features

	Embedded Voicemail	VoiceMail Lite	VoiceMail Pro
Runs as a service	No	No	Yes
Multi-lingual support	Yes	Yes	Yes
Voicemail for Individual users	Yes	Yes	Yes
Voicemail for Virtual users	Yes	Yes	Yes
Voicemail for Hunt Groups	Yes	Yes	Yes
Centralized Voicemail Services	No	No	Yes
Voicemail Ringback	Internal only	Internal only	Internal and external
Voicemail Help TUI	No	Yes	Yes
Message Waiting Indication	Yes	Yes	Yes
Visual Voice (interactive menu on phone display)	Yes	No	Yes
Integration with Phone Manager Pro	No	No	Yes
Personalized Greeting	Yes	Yes	Yes
Extended personal Greetings	No	No	Yes*
Continuous Loop Greeting	No	Yes	Yes
Forward to Email	Yes	Yes	Yes
Copy to Email	Yes	Yes	Yes
Listen To Email (Text To Speech)	No	No	Yes*
Send Email notification	Yes	Yes	Yes
Unified Messaging Service (UMS)	No	No	Option
Save Message	Yes	Yes	Yes
Delete Message	Yes	Yes	Yes
Forward Message to another Mailbox	Yes	Yes	Yes
Forward to Multiple Mailboxes	Yes	Yes	Yes
Forward with a Header Message	Yes	Yes	Yes
Repeat Message	Yes	Yes	Yes
Rewind Message	Yes	Yes	Yes
Fast Forward Message	Yes	Yes	Yes
Pause Message	No	No	Yes
Skip Message	Yes	Yes	Yes
LIFO/FIFO Message Playback Option	No	No	Yes
Set Message Priority	No	No	Yes*
Set automatic message deletion timeframe	No	No	Yes
Alphanumeric Data Collection	No	No	Yes*
Callers Caller ID, time and date announced	Yes	Yes	Yes
Call Back Sender (if Caller ID available)	Yes	Internal only	Yes
Remote Access to Mail Box	Yes**	Yes	Yes
User Definable PIN Code	Yes	Yes	Yes
Known Caller ID PIN Code By-Pass	Yes	Yes	Yes
Breakout to Reception	Internal and external.	Internal only	Internal and external.

- *Intuity mode only.
- **Remote access can be provided via the embedded Auto Attendant on the Small Office Edition.

© 2008 AVAYA All rights reserved.

In-Queue Announcements

	Embedded Voicemail	VoiceMail Lite	VoiceMail Pro
Queue Entry Announcement	Yes	Yes	Yes
Queue Update Announcement	Yes	Yes	Yes
Queue Position Announcement	No	No	Yes
Time in Queue Announcement	No	No	Yes
Time in System Announcement	No	No	Yes
Estimated Time to Answer (ETA)	No	No	Yes
Exit Queue to alternative answer point	No	No	Yes

Auto-Attendant/Audiotex

	Embedded Voicemail	VoiceMail Lite	VoiceMail Pro
Multi-Level Tree Structure	Yes	No	Yes
Message Announcements	No	No	Yes
Whisper Announce	No	No	Yes
Alarm Calls	No	No	Yes
Assisted Transfers	No	No	Yes

Other Features

	Embedded Voicemail	VoiceMail Lite	VoiceMail Pro
Call Recording	No	No	Yes
Test Conditions	No	No	Yes
Personal Numbering	No	No	Yes
Speaking Clock	No	No	Yes
Campaign Manager	No	No	Yes
VoiceMail Pro Manager	No	No	Yes
Customized Voicemail	No	No	Yes
Intuity TUI emulation mode.	No	No	Yes
Forward Emails to External Systems (VPIM)	No	No	Yes
Third Party Database Access (IVR)	No	No	Yes
Text To Speech within call flows	No	No	Yes
Support for Visual Basic Scripts	No	No	Yes

© 2008 AVAYA All rights reserved.

IP Office VoiceMail Pro Intuity Audix Emulation Features

Voicemail Box Feature	Intuity Feature support	VoiceMail Pro support
Basic Commands		
*4 (or *H)	Help	Yes
*7 (or *R)	Return to main menu	Yes
*9 (or *W)	Wait	Yes
**6 (or **N)	Look up number/name	Yes
**9 (or **X)	Exit system	Yes
0 or *0	Transfer call to operator	Yes
*3 (or *D)	Delete	Yes
**8 (or **U)	Un-delete	Yes
**4 (or **H)	Hold message in category	Yes
*8 (or *T)	Transfer out	Yes
**7 (or **R)	Log in again	Yes
Options while listening to messages		
9	Increase speed	Not supported
8	Decrease speed	Not supported
4	Increase volume	Not supported
7	Decrease volume	Not supported
6	Skip forward	Yes
5	Skip backwards	Yes
*6	Skip to next message component	Yes
*5	Skip to previous message component	Yes
2 or (*2)	Rewind to start of message (skip to previous message)	Yes
3	Play back header after pressing 2	Yes
*1	Print fax or text	Available as an option but fax messages not currently supported
Options for addressing voicemails		
*2 (or *A)	Alternate between name and number addressing	Yes
*5 (or *L)	Use mailing list for addressing	Yes
Responding to a message		
0	Call the sender	Yes, provided Caller ID is provided.
1	Reply to the sender by voicemail	Yes
2	Forward with comment at beginning	Yes
3	Forward with comment at the end	Yes
4	Record and address a message	Yes
Main Feature Support		
1	Record/Send messages	Yes
2	Get messages	Yes
3	Create greetings	Yes
4	Outgoing and filed messages	Not supported
5	Personal Options	Support for options 1, 3-7.
6	Outcalling	Yes.
7	Autoscan/Autoprint	Autoscan supported

© 2008 AVAYA All rights reserved.

PC Requirements

General Requirements

- An IP Office Feature Key is required for VoiceMail Pro.
- License for VoiceMail Pro and any additional ports required. If VoiceMail Pro server is installed without a license it will run for 2 hours and then shutdown.
- License for all options of VoiceMail Pro being installed.
- IP Office VoiceMail Pro CD.
- Installation on the same PC as being used for IP Office Manager is recommended.
- Switch off any PC and hard disk sleep, power down, suspend, hibernation modes.

PC Specification

- Always refer to the latest Avaya IP Office Technical Tip or Technical Bulletin for any updated information with regard to Operating Systems, Service Packs or PC hardware
- Refer to Technical Specifications section of the Product Description for Operating System and Hardware requirements

Network

- The Voicemail PC must be configured and tested for TCP/IP networking.
- The Voicemail PC must have a fixed IP address.

Disk Space

A compact or typical installation requires 500MB for the VoiceMail Pro software. A full installation requires up to 2GB of disk space. However prompts and recorded messages consume an additional 1MB of disk space per minute.

- For Avaya IP Office - Small Office Edition, you can expect to require at least 200 minutes of message recording space, that is 200MB.
- For a busy environment you can expect to require at least 1,000 minutes of message recording space, that is 1GB.

Web Server Operation

If web browser access to Unified Messaging Services (UMS) web service and or campaigns is required, Microsoft IIS Web Server must be installed on the Voicemail PC before VoiceMail Pro. Both applications must run as a service. A server operating system is required for IIS.

UMS Client Requirement

UMS IMAP access requires IMAP4 capable email clients such as Outlook. For voicemail web access see Web Server Operation above

Voicemail Email Connection

Voicemail Email operation is supported using either MAPI or SMTP. MAPI requires the VoiceMail Pro server PC to have a MAPI compliant email client install. See Voicemail Email Integration.

If Text to Speech is installed, email text to speech is supported using MAPI.

In both cases above, full email sending from the server PC to users PC should be configured and tested before VoiceMail Pro installation using the same PC user account under VoiceMail Pro will be installed.

IMS Pro Connection

IMS requires the Voicemail server to use MAPI.

- Integrated Messaging Pro (IMS) is supported on Microsoft Exchange 5.5, 2000 and 2003.

© 2008 AVAYA All rights reserved.

- An Exchange User account for user 'IMSAdmin' will be needed to as part of IMS installation.
- Must be a member of the same Domain as VoiceMail Pro Server.
- A list equating Exchange User account names with voicemail box users.

© 2008 AVAYA All rights reserved.

Voice Recording Library Management

IP Office Voice Recording Library (VRL) application is IP Office ContactStore. This application and its installation are documented separately. However:

- Avaya ContactStore for IP Office should be installed after VoiceMail Pro has been installed and its operation verified.
- Avaya ContactStore for IP Office must use a separate hard disk partition for its message archiving from that used by VoiceMail Pro for current mailbox messages. Use of a separate hard disk or installation onto a separate server PC are alternatives.
- The use of RAID 1 or RAID 5 are recommended.
- The use of a DVD recorder for long-term archiving is recommended.
- A figure of 7.2MB per hour of archived recordings is given.
- The archived messages held by IP Office ContactStore are accessed via web browser using the port address 8888. This port address is not configurable and so it is necessary to ensure that it does not conflict with any other web server service running on the same server PC.

© 2008 AVAYA All rights reserved.

© 2008 AVAYA All rights reserved.

Chapter 12.
Audio Conferencing

© 2008 AVAYA All rights reserved.

12. Audio Conferencing

A problem familiar to any organization is that of communicating effectively. As more and more people work from home or from dispersed locations, how do you ensure that employees are planning and working together effectively, and regularly keeping in touch when separated by time and distance? In addition, many companies choose to sub-contract some services such as payroll, logistics or manufacturing to third-party suppliers. How do you ensure that you can act as one virtual enterprise? Audio conferencing provides a simple and effective solution.

Audio conferencing makes it easy to include key people in decision making wherever they are with minimum interruption from their work. It responds to business needs that every company faces:

- More meetings but less time available.
- Increasing pressure to be at two locations at once.
- Travel restrictions.

As a result of using conferencing, the benefits gained are:

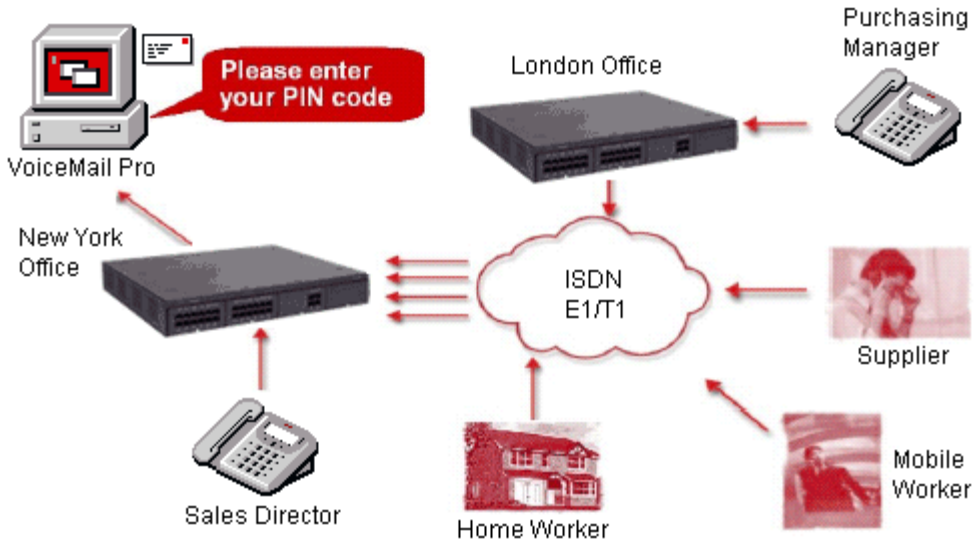
- Reduction in travel, leading to lower costs and less wasted time.
- Increased worker productivity and personal security.
- More effective working practices, leading to shorter project times, and supporting dispersed organizations and complex supply chains.

Furthermore, the Return On Investment (ROI) is very short as Meet Me conferencing is a built-in feature of IP Office. The typical ROI of just 4 to 6 months compared to Service Provider conferencing services based upon 2 hourly conferences with 5 participants per week.

© 2008 AVAYA All rights reserved.

IP Office Meet-Me Conferencing Solution

The conferencing solution built-in to IP Office enables multiple callers to talk in an audio conference. Callers can be on-site personnel as well as external parties whether field-based engineers, sales staff on the road, customers or suppliers. Conference calls can be planned in advance or established ad-hoc as and when required.



IP Office VoiceMail Pro complements the built-in meet-me conference bridge facility on IP Office systems by adding guidance prompts as well as requesting PIN codes as participants enter the conference for security. For example, if conference calls are regularly scheduled, VoiceMail Pro can have pre-programmed Call Flows for weekly conference calls e.g.: every Tuesday between 2pm and 5pm using PIN code 1234 for a sales call, etc. If multiple conference calls are scheduled, users can select which one they want to attend via a simple menu. Should users encounter any issues, calls can be automatically routed to the operator for assistance. For additional security, if Caller ID information is provided by the network VoiceMail Pro can make CallerID checks before allowing calls into a conference.

© 2008 AVAYA All rights reserved.

IP Office Conferencing Capacity

IP Office provides a flexible conferencing solution with from 3 to 64 way calling on an IP500 or IP406 V2 or 128 conference resources on an IP412. IP Office Small Office Edition provides 2 to 6 way calling with a maximum of 24 conference resources. This means that several conferences of different sizes can all run at the same time if the total calls do not exceed the systems conference resources. IP Office does not impose limits on the mix of internal and external calls in conference, but if all except one call disconnects from the conference bridge, the last calls is disconnected automatically by the system for added security.

Control Unit Conference Capabilities

The following tables show the maximum number of conference parties when calling via the different types of interface available on IP Office:

Maximum Participants	Small Office Edition	IP406 V2	IP412	IP500
E1 ISDN (Rest of World)	6	64	120	64
T1/PRI-T1	6	64/64	96/92	64/64
IP	6	30	60	64
Internal users	6	64	2x64	64
Total max.	**24**	**64**	**2x64**	**64**

Notes:

1. **Analog Trunk Restriction**
   In conferences that include external analog line calls, a maximum of two analog line calls are allowed per conference.
2. **External Participants**
   Each external caller requires a digital trunk/VoIP channel (for example 1 T1 allows 23/24 external parties, 1 E1 allows 30 parties and a VCM-20 allows 20 parties).
3. **Use of Conference Resources by Other Features**
   System features such as call intrusion, call recording and silent monitoring all use conference resources, as does automatic recording if enabled. When any of these features are active the number of slots available for conference parties is reduced.
4. **The IP412 Supports Two 64-party Conference Banks**
   When a new conference is started, the bank with the most-free capacity is used for that conference. However once a conference is started on one conference bank, that conference cannot use any free capacity from the other conference bank (i.e. no more than 64 parties in any one conference).
5. **Meet-Me Conferencing on IP500 requires Professional Edition**
   IP Office Standard Edition supports 64-way basic conferencing, but if Meet-Me capabilities are required the Upgrade License to IP Office Professional Edition should be purchased.
6. **IP Office Conferencing Center**
   If IP Office Conferencing Center is installed, 5 resources are reserved for use by the system. The maximum number of callers in any one conference and the total number of people on conference calls is reduced by 5. The maximum number of conferences on the system for IP406 V2, IP412 and IP Office 500 is reduced by 2.

© 2008 AVAYA All rights reserved.

IP Office Standard Conferencing Features

The IP Office provides the following features and benefits relating to conferencing:

- **No special conferencing equipment required**
  You only need an IP Office system unit with as many digital trunks/VoIP channels as external participants (as well as VoiceMail Pro should PIN code/menu prompts be required).
- **Ease of use**
  Simply dial the direct number allocated to the conference bridge, type in the PIN if required and you have joined the conference (PIN codes require VoiceMail Pro).
- **Conference control from IP Office Phone Manager Lite and Pro**
  For ad-hoc conferences with a few participants, users can easily set up immediate conferences by calling all parties and bringing them to the conference bridge. Thanks to IP Office Phone Manager, the instigator of the conference can keep control: the Caller ID number (and the associated name if recognized) of each participant is displayed within the Conference tab of Phone Manager. If required, he/she can selectively hang-up a specific participant.
- **Customized greeting**
  Record a personalized greeting per conference (requires VoiceMail Pro).
- **Conference entry/exit tones**
  Single beep on entry/double beep on exit
- **Conference call recording**
  Manual recording initiated by user on IP Office via Phone Manager, digital/IP display phone or a short code (requires VoiceMail Pro)
- **Security**
  To prevent unauthorized access to the conference bridge, PIN codes, Caller ID number screening as well as time and date profiles can be set-up using IP Office VoiceMail Pro.
- **Privacy**
  In cases where the security of calls is critical, in-house conferencing is the only way to ensure privacy.
- **Remote Management**
  Allows a single person to manage the conferencing bridge facility from any location. Furthermore, the full IP Office solution - phone system, voicemail, CTI server, router, firewall and DHCP server- can all be managed from a single management interface called IP Office Manager.

© 2008 AVAYA All rights reserved.

Conferencing Center

The integrated conferencing functionality on IP Office is enhanced by adding Conferencing Center. This optional licensed application is a web-based software package that consists in two parts:

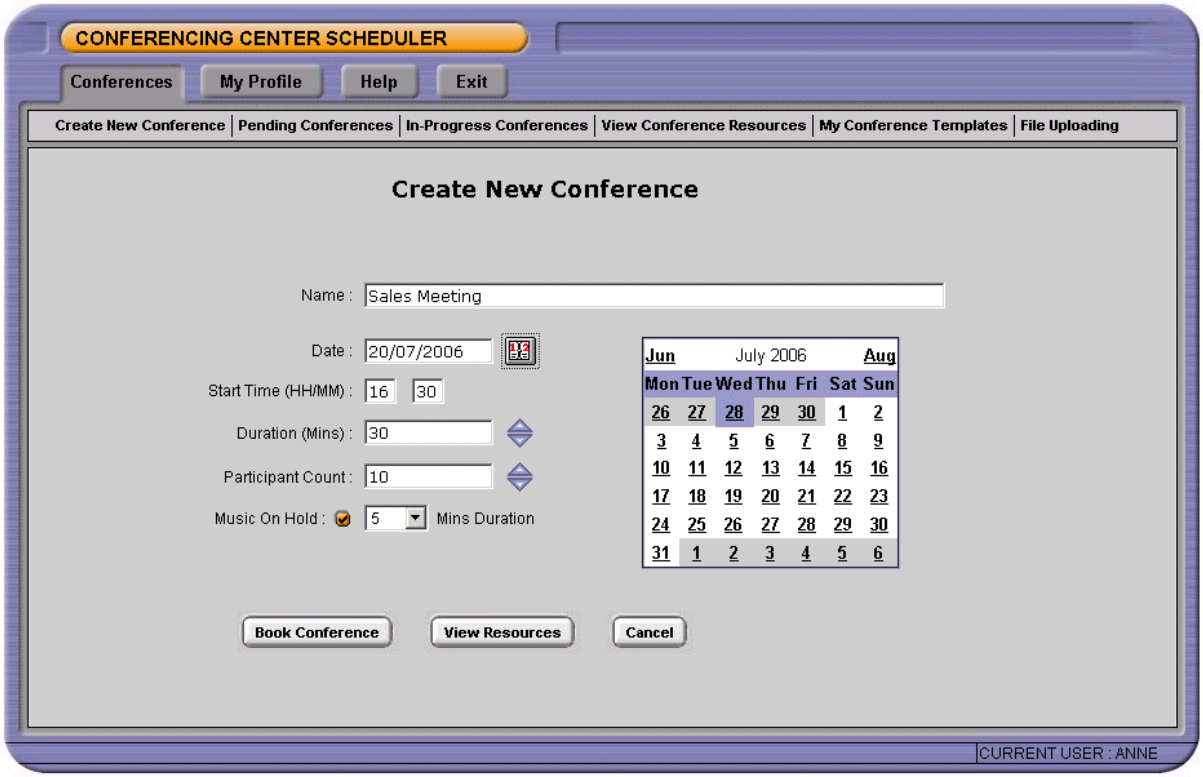
- a "Conferencing Center Scheduler" to book and reserve conferences.
- a "Conferencing Center web client" to complement an audio conference with a visual presentation web interface.

The scheduler is independent of the web client, either or both can be used. Conferencing Center also interacts with SoftConsole and Phone Manager.

Note: Conferencing Center on the IP500 requires a license for IP Office Professional Edition.

Conferencing Center Scheduler

The Web Scheduler allows registered users to create and book conferences online using a web client interface. The Scheduler offers secure conferencing while being very easy to set up. Users simply enter the date, time, duration and the number of conference participants required. The conference is created, if the resources are available for that specific time. Once reserved, the conference resources are allocated to that conference call for the specified number of participants at the selected date and time. Additionally Music On Hold (if available on the system) can be played to callers while waiting for the conference to start.



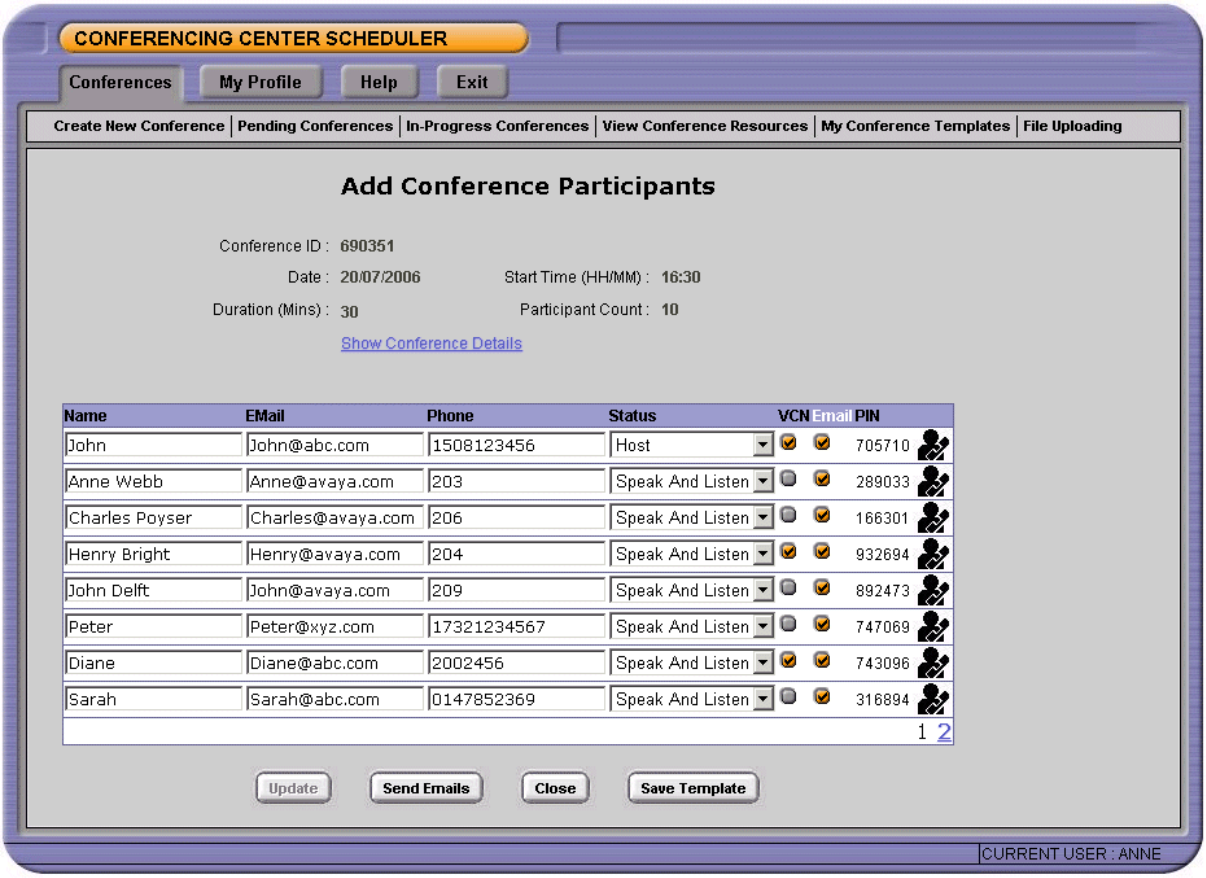
Access to the Web Scheduler requires a user to be granted a user logon and password by the administrator and have Internet Explorer (6.0 or above) installed on their PC. No other software is required. The System Administrator can set up an unlimited number of registered users on the Conferencing Center application. Once registered, users can review the system resources before booking a new conference, book a conference as well as list pending conferences they have previously set up.

© 2008 AVAYA All rights reserved.

The user setting up the conference can then add participant details including their email address and their telephone number. This allows email notification to all participants confirming the conference call details including the conference name, description, host contact details, bridge number, conference ID, their unique participant PIN code (if PIN checking has been selected) and the URL web address for the web client (if web support has been selected). At any time prior to the start of the conference, Participants' details can be changed.

Voice Conferencing Notification (VCN) can be activated for selected participants. This allows VoiceMail Pro to dial out to participants when the conference is about to start and bring them to the conference bridge if they are available.

Advanced security is available by generating unique PIN numbers for every participant allowing them to be recognized by the system and displayed on the Conferencing Center Web client (if selected – see paragraph below). If caller announcements are required, VoiceMail Pro can announce each participant by asking them for their name which is then announced to all participants already on the bridge. Similarly at the end of the conference, each participant leaving the conference will be announced.

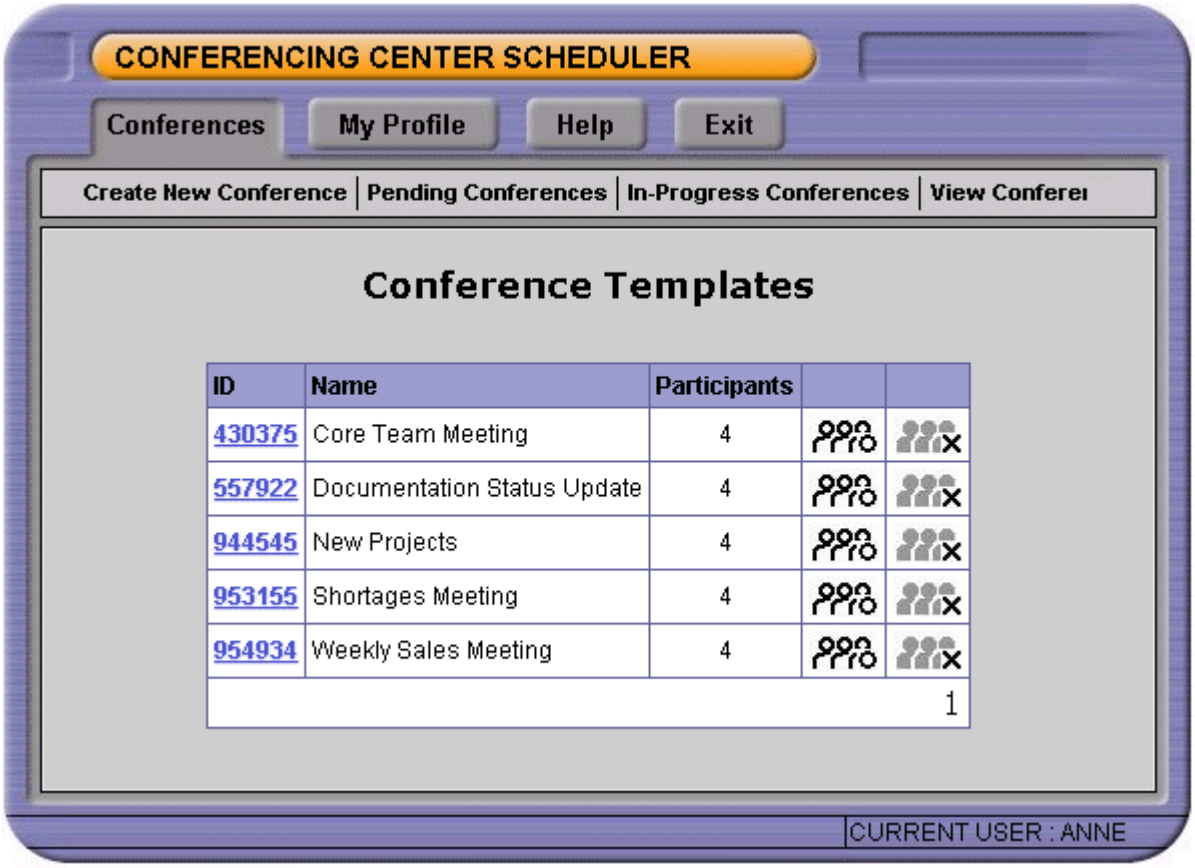

CONFERENCING CENTER SCHEDULER
Conferences
My Profile
Help
Exit
Create New Conference | Pending Conferences | In-Progress Conferences | View Conference Resources | My Conference Templates | File Uploading
Add Conference Participants
Conference ID : 690351
Date : 20/07/2006
Start Time (HH/MM) : 16:30
Duration (Mins) : 30
Participant Count : 10
Show Conference Details
Name EMail Phone Status VCN Email PIN
John John@abc.com 1508123456 Host 705710
Anne Webb Anne@avaya.com 203 Speak And Listen 289033
Charles Poyser Charles@avaya.com 206 Speak And Listen 166301
Henry Bright Henry@avaya.com 204 Speak And Listen 932694
John Delft John@avaya.com 209 Speak And Listen 892473
Peter Peter@xyz.com 17321234567 Speak And Listen 747069
Diane Diane@abc.com 2002456 Speak And Listen 743096
Sarah Sarah@abc.com 0147852369 Speak And Listen 316894
1 2
Update
Send Emails
Close
Save Template
CURRENT USER : ANNE

© 2008 AVAYA All rights reserved.

A local address book facility is available to provide a convenient method of managing conference contacts and using these contacts when booking a conference. The address book can be accessed in two ways, either from the 'My Profile' tab or from the Add/Update Conference Participants process.

© 2008 AVAYA All rights reserved.

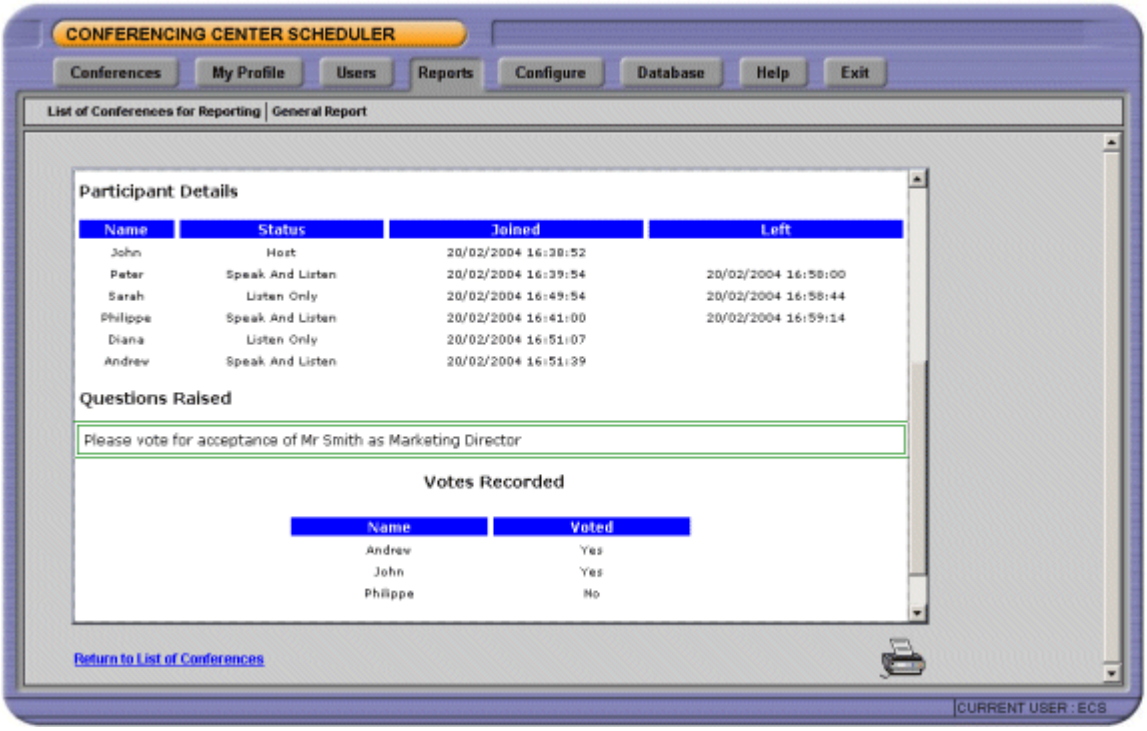
Conference templates can be used to book recurring conferences, all booking information including the conference ID and participants PINs are retained, except for the conference date. Using a conference template in this way can save re-entering of repetitive information thus saving time and effort. Once a template has been created they can be accessed via the 'My Conference Template' tab:



© 2008 AVAYA All rights reserved.

Conferencing Center Reporting

The System Administrator can generate reports regarding conference usage and individual conference reports. This will detail the conference name and ID, the start date and time, duration and number of participants. If PIN codes were used, individual reports can be run listing participant details and when they joined/left the conference. Finally, if voting was being used using the Conferencing Center Web Client, voting results for each participant would be shown for each question asked during the conference call.



In summary, the Conferencing Center Web Scheduler offers the following:

- Web-based booking tool to reserve conference resources (immediate or future).
- Ability to select "Listen-only" or "Speak and Listen" mode for each participant.
- Email notification to all participants.
- Voice Conference Notification (VCN) to dial out participants.
- Participants name announcements as they enter/leave the conference bridge.
- Unique computer-generated Conference ID for security.
- Unique PIN code for each participant for security and authentication.
- Web-based reports on conference usage and voting results.

© 2008 AVAYA All rights reserved.

Conferencing Center Web Client

To complement the audio-conference, the host has the ability to share information over the Internet. The Web Client offers a browser interface where the host and participants can not only see which participants have joined the conference but also whether they joined as audio-only or both audio and web. A conference host has the ability to pose questions, modify participant speak/listen settings and whisper to a single participant connected into the conference. When in listen-only mode, participants can request the right to speak through their Web Client (raise hand function). A Web Chat service is available between Host and Participants and the dialog is recorded and sent via email to the Host after the conference. Two modes of communication between Host and Participant is supported, either private or public. Public allows all participants to see the dialog

The host can present a document on the Web Client with all participants. (for example a PowerPoint presentation, Word document or an Excel spreadsheet) or simply a website URL. Files can be loaded on demand using the Web Client, or in advance using the Web Scheduler. When presenting the document, the host has the ability to synchronize the document view to all participants (e.g. change slide) as long as he resides within the same IP domain as the Conferencing Center server (this is a Microsoft limitation).

Participants can be located anywhere on the Internet or across an extranet as long as they have access to the Web Server running the Conferencing Center application.

Access to the Conferencing Center Web Client requires the participant to have Internet Explorer (6.0 or above) installed on their PC. No download of the application is required. There can be as many web clients as there are participants on the conference call (up to 64 maximum per conference). For security, access to the Web Client requires the participant to logon using the Conference ID and their unique PIN number. This allows the system to recognize who joined the conference and display its name on the right-hand side of the screen.

In summary, the Conferencing Center Web Client offers the following:

- Real-time view of participant's status (Dialed in, Logged on to Web client, Speak and Listen, Listen Only).
- Ability for the host to change participant status in real-time.
- Ability for participants in listen-only mode to request the right to speak (raise hand function).
- Mute All / Un-Mute All facility for the host.
- Web Chat between Host and Participant
- Whisper facility for the host to have a private conversation with one of the participants.
- Viewing area for reviewing PowerPoint presentations, Word documents and Excel spreadsheets.
- Questions and Voting facility.

© 2008 AVAYA All rights reserved.

Phone Manager Conferencing Center Integration

Phone Manager users can join a conference or book a conference via the Conferencing Center application by clicking the relevant icons within Phone Manager. This will launch the Conferencing Center Web Client and the Conferencing Center Scheduler respectively. Note this feature is only available if permission is specified by the system administrator and if the Conferencing Center system is installed and available.

System Requirements for Conferencing Center

Conferencing Center Server PC Specification

- Always refer to the latest Avaya SMB Technical Tip or Technical Bulletin for any updated information with regard to operating systems, service packs or PC hardware.
- Refer to the Technical Specifications section of the Product Description for operating system and hardware requirements.

Conferencing Center web client:

- Internet Explorer 6.0 or higher.
- No download required.

© 2008 AVAYA All rights reserved.

© 2008 AVAYA All rights reserved.

Chapter 13.
The Contact Center

© 2008 AVAYA All rights reserved.

13. The Contact Center

Avaya provides Customer Contact solutions that meet the needs of the small to medium business. From the smallest company that requires basic system performance reporting to the larger businesses that need advanced routing and multimedia integration with the Customer Contact Center. Avaya provides two solutions on the IP Office communications platform based on the needs of Small Business Customers:

- **Small Business Contact Center** 242 *NEW*
- **Compact Contact Center** 251

© 2008 AVAYA All rights reserved.

Small Business Customer Service Overview

Contact Centers are a requirement at nearly every level of business, from very small informal groups of 5 or 10 "customer service reps", to large formal campaigns that can involve thousands of agents.

However, customers in the small business market are generally very nervous about making such a high profile investment in a "contact center solution." They require a "business intelligence solution" that deploys easily, can be managed with very little training, and can provide statistics on each segment of the business. Small Businesses also do not have the resources for extra IT staff to administer databases, servers, etc, so they need a solution that can be easily understood by their own employees. It should operate as a service to each of the clients.

SBCC Concept

SBCC is Avaya's new server based contact center product designed explicitly for small businesses. Drawing upon the latest web and design technologies the SBCC introduces significant new capabilities to effectively and efficiently manage a multi-site call center environment. The product focus is on ease of use and deployment simplicity to ensure improved ongoing productivity and lower support costs.

In its native web server environment the SBCC can be simply installed onto a single server without the need for any client software deployment. Administration and management of the call center is carried out via a thin client through a secure password protected web browser session.

In recognition of the need to monitor and report upon agents in disparate locations the SBCC now includes the capability to provide coverage of call center activity across a Small Community Network (SCN). A suitably licensed IP Office node in a SCN network can now be monitored in real time and reported upon by way of one of the seven new reporting templates that are included as part of a simple yet powerful reporting package.

SBCC Key Features

- Real time Call Center activity monitor and historical reporting across a SCN network.
- Microsoft server based with thin client (web browser) client connections.
- Simplified installation and maintenance.
- Seven reporting templates provide drag and drop and user defined filter functionality.
- Three customizable Supervisor views and summarized Agent view.
- Report Scheduler.
- Real time scrolling alarm bar.
- Multi-lingual capable.
- Supports up to 30 Supervisors and 150 Agents.
- IP Office 4.2+ platform support.
- Removal of CCC highest numbered group rule.

SBCC Business Benefits

- **Lower TCO**
  SBCC provides small businesses with a web based contact center measurement tool, producing cradle to grave reporting in an easily understandable format with no client software to load.
- **Standards Based**
  SBCC uses standards based applications, such as Microsoft SQL 2005 Express, and supports all major web browser software, to provide small business installers with greater flexibility in deployment.
- **Ease of Use**
  SBCC's real-time charts can be customized by the user in the manner that suits their business best, with historical reporting templates that allow the business to filter on the type of date they want to see.

© 2008 AVAYA All rights reserved.

Ordering Small Business Contact Center

Small Business Contact Center supports the following maximum configurations for IP Office:

- 150 Agents
- 30 Supervisors

Each IP Office, used in a Small Community Networking environment, which is part of the SBCC, must hold sufficient licenses for their local Agents.

SBCC Sales Order Codes

SAP Code	Description	Notes
217650	IPO LIC SBCC 1AGT RFA	Agent packages come in 1, 5, 10 and 20
217651	IPO LIC SBCC 5AGT RFA	
217652	IPO LIC SBCC 10AGT RFA	
217653	IPO LIC SBCC 20AGT RFA	
217654	IPO LIC SBCC 50AGT RFA	
217655	IPO LIC SBCC 1SPV RFA	Supervisor licenses for Real Time and Historical Reporting
217656	IPO LIC SBCC 10SPV RFA	
217657	IPO LIC SBCC 20SPV RFA	
217658	IPO LIC CCC-SBCC UPG PCKG	License that upgrades existing CCC licenses to SBCC for CCC users (no database upgrade).
700461874	IPO DVD SBCC	DVD Containing required software for SBCC

To activate an SBCC system all that is required is the purchase of one of the supervisor licenses (1, 10, 20) and the system will be able to report on any agent that has been licensed properly. Licensing is on a concurrent basis, i.e., once an agent logs out of SBCC, that license becomes freed up for the next agent to use. This is effective for those businesses that are working in shifts.

IP Office 4.2 will have a setting in IP Office Manager for designating which hunt groups will be reported upon by SBCC. Any hunt group that contains agent who is not properly licensed will not be able to retrieve any reporting on that group (real-time or historical).

SBCC Server Operating System Requirements

Small Business Contact Center server will be supported on the following 32 bit platforms only:

- Microsoft Windows Business Server 2003 R2.
- Microsoft Windows Server 2003 SP2.
- Microsoft Windows Server 2003 R2.

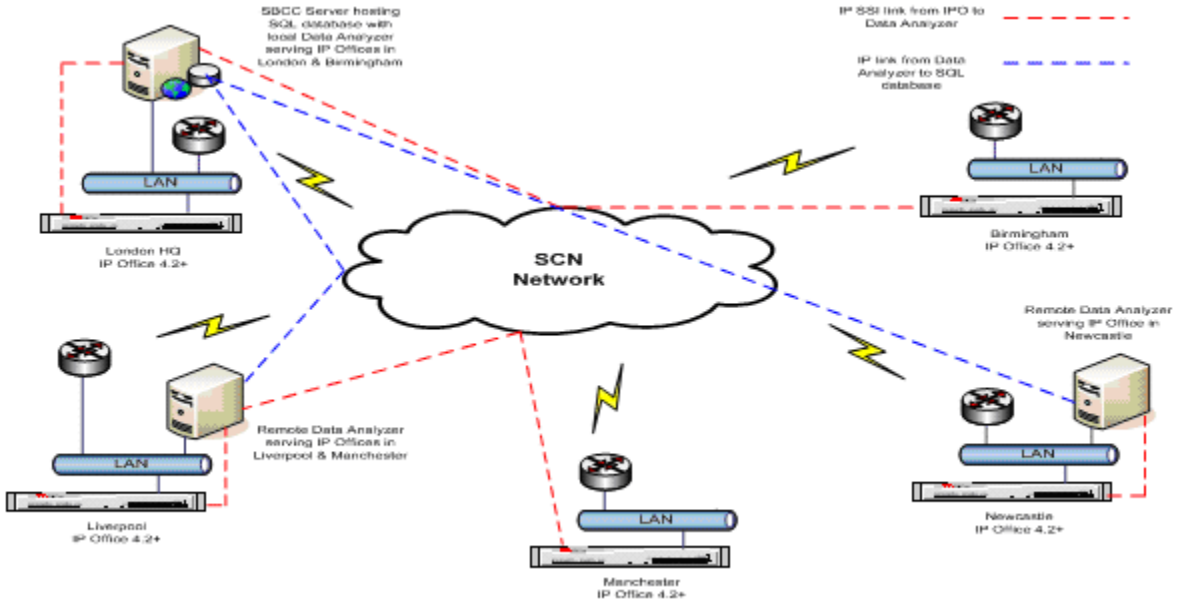
Support for Windows 2008 server (both 32 and 64 bit) is planned for a future release.

© 2008 AVAYA All rights reserved.

SBCC Server Hardware Requirements

The minimum server specifications are as follows:

- **SBCC server**
  Intel Pentium D 945 core/AMD Athlon 64 4000+ ; with 2GB RAM and 30 GB free hard disk space. Note: The SBCC server can co-reside with a VoiceMail Pro of less than 17 ports.
  - **Stand Alone Data Analyzer**
    Intel Pentium D 945 core/AMD Athlon 64 4000+ ; with 1GB RAM and 1 GB free hard disk space
    - One Data Analyzer can support a maximum of two IP Office sites where licensed SBCC agents exist. For example, an IP Office in a Small Community networking arrangement with 4 sites would require two servers; one SBCC server for the primary site, and one node, and one stand-alone data analyzer to serve the remaining two sites. Please refer to the diagram below


SBCC Server hosting SQL database with local Data Analyzer serving IP Offices in London & Birmingham
IP SSI link from IPO to Data Analyzer
IP link from Data Analyzer to SQL database
LAN
London HQ IP Office 4.2+
SCN Network
LAN
Birmingham IP Office 4.2+
Remote Data Analyzer serving IP Offices in Liverpool & Manchester
LAN
Liverpool IP Office 4.2+
LAN
Manchester IP Office 4.2+
Remote Data Analyzer serving IP Office in Newcastle
LAN
Newcastle IP Office 4.2+

SBCC Client Components

The client application consists of a secure thin client web browser session to the SBCC server that will provide the Supervisor or Agent with a complete view of the Call Center.

The following web browsers will be supported as thin clients on a Windows OS:

- Microsoft Internet Explorer, version 7.0 and above.
- Mozilla Firefox, version 1.5 and above.
- Opera, version 9 and above.

SBCC Database

The database has been designed to provide a simple table schema. The database is a Microsoft SQL Server 2005 instance and the SBCC will be packaged with the Microsoft SQL Server 2005 Express edition. It is recommended that any call center that predicts high call usage should understand the imposed limitations of Microsoft SQL 2005 Express edition and consider using the full version of Microsoft SQL 2005. Microsoft SQL 2000 and 2008 are not supported.

The database is a "pre-requisite" that must be installed prior to the SBCC install. During installation the default location for the installation of the database is on the local drive. This can be changed to a remote location on the network. If an existing database is to be used then the option for this to exist on a remote server also applies.

© 2008 AVAYA All rights reserved.

SBCC Historical Reporting

The SBCC reporting package has been simplified to enable the user to easily retrieve the reporting information they need from one of seven standard drag and drop report templates. The SBCC offers "cradle to grave" reporting for businesses that want to track specifically how a call was routed through their business. The historical reporting is accessed via the Supervisor login and corresponding view as shown below.

SBCC Report Templates

The following main report templates are available with SBCC:

- **Call Details Report**
  The Call Details Report contains details of the total number of calls, internal calls (optional), external outbound calls and the total number of external inbound calls.
- **Call Summary Report**
  The Call Summary Report contains details of the total calls presented and answered; the average speed to answer percentage and average speed to answer time, the number of calls overflowed and lost, the average abandon time as well as the grade of service.
- **Customer Service Summary**
  The Customer Service Summary Report will contain details of the total logged in time, total ringing, talk outbound, talk inbound, talk internal, total pause, total ACW, total holding, including parked calls and total off hook.
- **Trace Reports**
  The Trace Report lists in chronological order all the events, date and time stamped that involved the subject selected.
- **Alarm Reports**
  Alarm reports can be created for the Supervisor Views. All the alarms for the selected subject are listed over a user specified time period. Totals are included that show the number of warning and critical alarms received for each unique alarm type included in the report.
- **Graph Reports**
  There are 9 types of graphs that can be produced for specific criteria. All the graphs can be produced or just a specific graph. When more than one graph is selected, two graphs are produced per page of output. The graphs are volume (Y axis) plotted against the selected time period (x axis). The following types of graph reports are available; Agents Busy, Average Speed of Answer (ASA), Grade of Service (GOS), Lost Calls, Trunk Capacity, Calls Waiting, and Agent Availability.
- **Voicemail Reports**
  The report shows the following information; Total calls answered by the Auto-Attendant (VMPro), total calls abandoned (lost) in Auto-Attendant, and a table of unique call flow trigger names with a tally of the number of times the trigger has been received (e.g. who pressed 1 for sales in the Auto-Attendant).

© 2008 AVAYA All rights reserved.

Creating an SBCC Report

From the seven basic report templates within SBCC, a user can create over 100 different reports depending upon how they want to see the information that is most useful for their business. All standard report templates have common fields that can be completed to create a report. And when entering details in a field a wildcard '* ' can be used to represent everything. The common fields are listed below:

- **Report Name**
  The text entered is used as the main title of the report. If the report is saved this is the saved name of the report.
- **Subject**
  This is the main subject of the report. The drop down list offers a list of report subjects that are pertinent to the report i.e. Hunt Group, Agent, DDI etc.
- **Filter**
  The filter is used to refine the data and again offers filters that are pertinent to the report i.e. All, Answered, Lost, etc.
- **Exclude Internal**
  When selected, internal calls are excluded from the report.
- **Working Hours**
  The start and finish time of the working day can be specified using the spin controls.
- **Group Results by**
  The drop down list offers a list of display options.
- **Scheduled**
  By default this field is shown as 'Manual'. If the option 'Scheduled' is selected then additional fields are available:
- **Report Format**
  The format that the report is to be produced in can be selected from Adobe, Word, Excel, Rich Text or Crystal.
- **Email Address**
  An email address can be entered if the report is to be sent by email.
- **Printer**
  The printer path that the report should be sent to.
- **Report Period**
  Selection of days, weeks or months from a list box. The number of days or weeks is counted from the date that the report is first scheduled to be run. Months are from the last full months data i.e. if the report is scheduled in February then January will be the last month in the report.
- **Report Range**
  The start and end date for the report.
- **Language**
  If the default language is not required then another language can be selected that your system supports.

Filters can be used in the report to provide exactly the type of information the customer is looking for. The filter field is used to refine the data and offers the following options in a drop down list; All, Answered, Refused, Refused Overflowed, Overflowed All, Overflowed Off PBX, Overflowed on PBX, Transferred and Lost Calls. The default setting is all.

Reports can also be scheduled for future delivery to any network printer or email address in any of the supported formats.

© 2008 AVAYA All rights reserved.

SBCC Report Examples

Call Details Report - Filters: All Agents/Answered

Answered - All Agents

From 20/03/2008 To 28/03/2008
Working Hours 09:00 To 17:00

Report type	Target Name	Supervisor Name
Calls Details Report - Answered	Hunt Group	Chris Boseley

All Agents

Date - Time	Calls Direction	CLI	Queue	Queued Time	Agents	Call Duration	Status	Reference
27/03/2008 16:21:46	Internal	01707392200	All Agents		Stephen	00:00:10	Answered	31
27/03/2008 16:22:01	Internal	01707392200	All Agents		Sterling	00:00:10	Answered	32
27/03/2008 16:22:16	Internal	01707392200	All Agents		Steve	00:00:10	Answered	33
27/03/2008 16:22:20	Incoming		All Agents		Steven	00:00:17	Answered	35
27/03/2008 16:22:31	Internal		All Agents		Stew	00:00:10	Answered	36
27/03/2008 16:22:46	Internal		All Agents		Susan	00:00:10	Answered	39
27/03/2008 16:22:45	Incoming		All Agents		Sue	00:00:17	Answered	38
27/03/2008 16:23:01	Internal		All Agents		Teresa	00:00:10	Answered	40
27/03/2008 16:23:16	Internal		All Agents		Terry	00:00:10	Answered	43
27/03/2008 16:23:10	Incoming		All Agents		Terri	00:00:17	Answered	42
27/03/2008 16:23:31	Internal		All Agents		Thad	00:00:10	Answered	44
27/03/2008 16:23:36	Incoming		All Agents		Thomas	00:00:16	Answered	46
27/03/2008 16:23:46	Internal		All Agents		Thomy	00:00:10	Answered	47
27/03/2008 16:24:01	Internal		All Agents		Timothy	00:00:10	Answered	50
27/03/2008 16:24:01	Incoming		All Agents		Tiffany	00:00:16	Answered	49
27/03/2008 16:24:26	Incoming		All Agents		Tisha	00:00:16	Answered	53
27/03/2008 16:24:51	Incoming		All Agents		Todd	00:00:16	Answered	55
27/03/2008 16:25:16	Incoming		All Agents		Tracey	00:00:16	Answered	57
27/03/2008 16:25:41	Incoming		All Agents		Troy	00:00:17	Answered	59
27/03/2008 16:26:06	Incoming		All Agents		Vaughn	00:00:17	Answered	61

No. Of Calls	20	No. Of Ext Inbound Calls	10
No. Of Internal Calls	10	No. Of External Outbound Calls	0

Collective CW

Date - Time	Calls Direction	CLI	Queue	Queued Time	Agents	Call Duration	Status	Reference
27/03/2008 16:35:42	Internal		Collective CW		Stephen	00:00:10	Answered	130
27/03/2008 16:35:57	Internal		Collective CW		Stephen	00:00:10	Answered	135
27/03/2008 16:36:12	Internal		Collective CW		Stephen	00:00:10	Answered	141
27/03/2008 16:36:27	Internal		Collective CW		Stephen	00:00:10	Answered	148
27/03/2008 16:36:42	Internal		Collective CW		Stephen	00:00:10	Answered	155

Call Summary Report - Filter: All Agents

All Agent Hunt Group Summary

From 20/03/2008 To 28/03/2008
Working Hours 09:00 To 17:00

Report Type	Target Name	Supervisor Name
Calls Summary Report - All	Hunt Group	devesh

Hunt Group	Calls Presented	Calls Answered	Avg. Speed To Answer (%)	Avg Speed To Answer (Time)	Calls Overflowed	Calls Lost	Avg. Abandon Time	Grade Of Service (%)
All Agents	20	20	100.00	00:00:04				100.00
Collective CW	20	20	100.00	00:00:05				100.00

Total Presented Calls	40	Avg. ASA (%)	100
Total Answered Calls	40	Avg. ASA Time	00:00:04
Total Overflowed Calls		Summary Abandon Time	
Total Lost Calls		Grade of Service	100

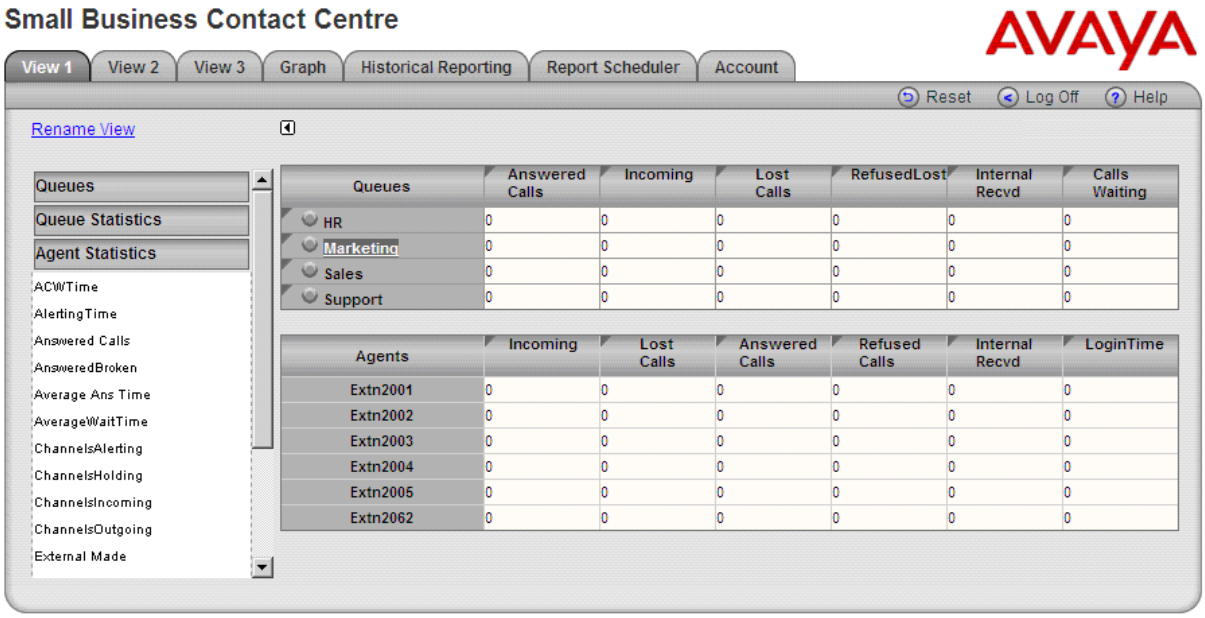
© 2008 AVAYA All rights reserved.

SBCC Real Time Reporting

Small Business Contact Center provides real time monitoring of call center activity. This is achieved via a thin client web browser session logged onto the SBCC server. Within the supervisor client a user can program up to 3 different views of their call center. This is a very useful feature as now a supervisor can divide their responsibilities into different views and compare the results of those views.

For example, assume that a call center supervisor is responsible for the following hunt groups in IP Office; Computer Sales (201), Monitor Sales (202), Warranty Service (203), and non-Warranty service. The Supervisor can create two separate views entitled "sales" and "service" and compare the number of calls, length of call, etc. This allows a supervisor to see a whole system as it relates to the business function (sales, service).

The Supervisor views are private and can only be seen with a login. A view may be created that shows specific Hunt Group and Agent statistics as well as Alarms and Warnings. Each of the three views are summarized and made available to any Agents when they logon and select their corresponding Supervisor. An example of the Supervisor real-time view is shown below:



© 2008 AVAYA All rights reserved.

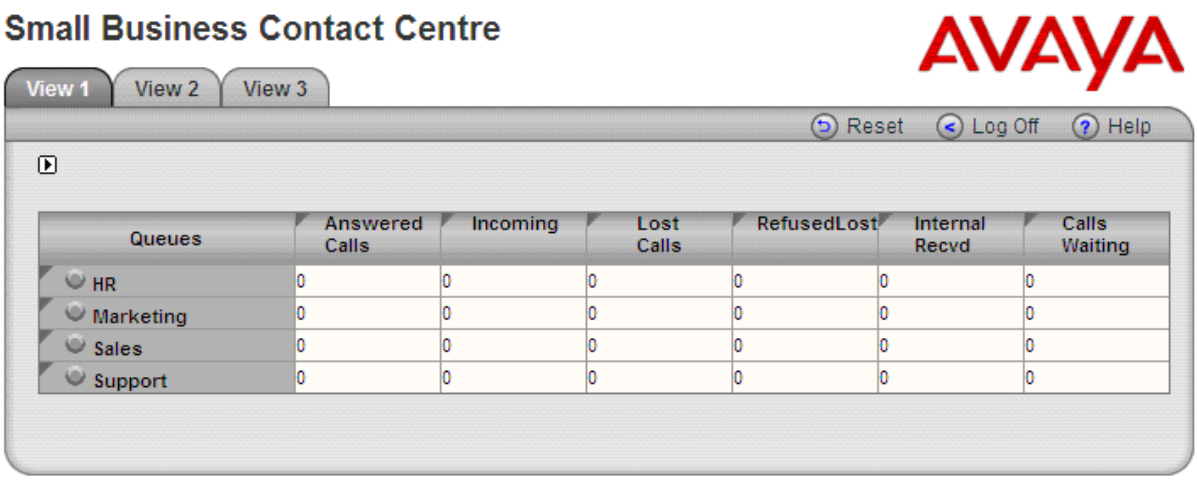
Alarms

The Small Business Contact provides different levels of alarms for the supervisor and agent to properly manage their call center activity. As parameters move from a normal to alarm state, the color of the field within the realtime view will change depending upon the level of alarm; white for normal, yellow for caution, red for alarm, and blue for alarm acknowledged. The following statistics can have alarms set against them within the SBCC:

- Answered Calls
- Answered Internal (Direct)
- Answered Internal (Hunt Group)
- Answered External (Direct)
- Average Speed of Answer Percentage (ASA%)
- Average Speed to Answer Time
- Call Share
- Calls Waiting
- Agent State and Time in State (System)
- Agent State and Time in State (This Group)
- Grade of Service
- Hunt Group State and Time in State
- Internal Made
- Lost Calls
- New Messages
- Num Agents After Call Work
- Num Agents Available
- Num Agents Logged On
- Num Agents Pause Time
- Num Agents Present
- Num Agents Ringing
- Outbound Calls
- Overflowed Calls
- Refused Calls

Agent View (PC Wallboard)

In Small Business Contact Center, the Agent view is part of the agent reporting license, every licensed agent will be able to run their own view. An Agent is able to login via a thin client web browser session and associate with a Supervisor. The Agent view also displays any alarms that have been set by the Supervisor and is featured to act as a PC Wallboard display.



© 2008 AVAYA All rights reserved.

SBCC Call Statistics

The Small Business Contact Center provides detailed statistics to help small businesses understand exactly where a call has been within the business. Statistics are measured against (also refer to the table below):

- **Queues**
  Points in the call center where a queue forms such as Hunt Groups.
- **Agents**
  Individuals that are members of queue groups who answer the calls.

Statistic	Queue	Agent
Answered Calls	✓	✓
Answered External (Direct)	✗	✓
Answered Internal (Direct)	✓	✓
Answered Internal (queue)	✓	✓
Average Speed of Answer Percentage (ASA%)	✓	✓
Average Speed to Answer Time	✓	✓
Call Share	✓	✓
Calls Waiting	✓	✗
Agent State and Time in State (System)	✗	✓
Agent State and Time in State (This Group)	✗	✓
Grade of Service	✓	✗
Queue State and Time in State	✗	✗
Internal Made	✗	✗
Lost Calls	✓	✓
New Messages	✓	✓
Num Agents After Call Work	✓	✗
Num Agents Available	✓	✗
Num Agents Logged On	✓	✗
Num Agents Pause Time	✓	✗
Num Agents Present	✓	✗
Num Agents Ringing	✓	✗
Outbound Calls (External)	✗	✓
Overflowed Answered	✓	✗
Overflowed Calls	✓	✗
Overflowed Calls Waiting	✓	✗
Overflowed Lost	✓	✗
Refused Calls	✓	✓
Transferred	✓	✓

© 2008 AVAYA All rights reserved.

Compact Contact Center

IP Office Compact Contact Center is a modular contact center solution catering for all contact center sizes from 2 to 75 agents. The following modules are available as part of the CCC software application:

- **Compact Call Center (CCC) Server - Base System**
  Provides one supervisor position with real-time information view, management by exception, and historical reports for any aspect of the contact center. Up to 73 standard reports can viewed or printed. Also included are reporting capabilities on 5 agents and one license for a PC Wallboard (PCWB) application.
- **Agent and Site Management (Real Time)**
  - **Real Time Supervisor Monitoring - Call Center View**
    As many as 21 supervisor CCV positions can be used in CCC (please note: MSDE installations can only be supported up to 5 supervisor positions). This provides a supervisor with the ability to monitor in real time the service being provided to callers. There are up to 12 separate real-time graphs that can be viewed by the supervisor. Alarms also appear in real time prompting the supervisor to acknowledge them as they occur.
- **Phone Manager Pro: Agent Enabled**
  Provides agents with a PC CTI application where they can log in, join groups, and go into busy status when they are unable to accept calls for short periods so no special turrets are needed – CCC and Phone Manager allow Agent working on any wired IP Office extension type. Phone Manager PC Softphone can be used in agent mode as well, without the need for a physical telephone. Please refer to the applications section for more information on Phone Manager Pro.
- **Alarm Reporter**
  Alarm Reporter is designed to enhance the exception management used by Call Center View (CCV). The Alarm Reporter enables the contact center supervisor to look back on the performance of the contact center, on a daily or weekly basis, by reporting on certain criteria predefined by the contact center supervisor.
- **Historical Reporting**
  The Compact Contact Center archives all call center interactions (telephony or multimedia) to a central database (MSDE or SQL). This provides the data source for a set of standard reports to the business, and the capability to create custom reports.
- **CCC Reporter**
  The system allows up to 20 separate Report Viewers within the contact center (for MSDE installations, up to 5 viewers are supported). Access to the standard reports is a thin client application based on Crystal Reports. Up to 73 standard reports are available, with the ability to create 3 more custom reports, see custom reports section below. Reports can be exported to a variety of formats, including Excel, CSV, HTML, and PDF.
- **Report Scheduler**
  All historical reports created within CCC can be scheduled for individual delivery to anyone via email or sent to multiple network printers.
- **Custom Reports**
  All CCC reports are created through Crystal Reports. This application provides a much richer experience for the small to mid-market customer, and creates an environment where custom reporting is more accessible. To create more than 3 CCC custom reports requires the designer license (IPO CCC DESIGNER RFA) AND a compatible version of Crystal Reporting software (Crystal version 9).
- **Wallboards**
  - **Fixed Wallboards**
    Fixed scrolling wallboards enable key statistics and messages to be displayed for everyone in the contact center to see. Supervisors can send ad-hoc messages to wallboards to broadcast important information, or to make announcements.
  - **PC Wallboards**
    PC-based wallboards allow individual agents to see their own individual statistics, those for their group, or for the whole contact center. Agents can customize their view so that information is presented in the way most useful to them. In additional, supervisors can set particular messages to appear on PC Wallboards, as a motivational or informational tool. Please refer to the CCC System Administration manual for a complete list of variables available.
- **3rd Party Integration**
  - **Microsoft TAPI Integration**
    By utilizing either the 1st party or 3rd party TAPI support on IP Office businesses can link their contact management to their telephony (e.g. ACT! Goldmine) and increase the productivity of their agents and the profitability of the contact center.

© 2008 AVAYA All rights reserved.

Call Center View - Real Time Reporting

Supervisors in a contact center are there to manage workload. Call Center View provides the Supervisors with the combination of real time service monitoring and resource management, allowing them to balance and manage their resources (i.e. staffing levels against the traffic levels of incoming calls) and therefore improve customer service and reduce costs. Call Center View contains 18 real time screens showing all aspects of the Contact Center activity. Alarms may be set on up to 16 parameters per device, with three levels per alarms available, ensuring that a supervisor will be informed should an exception occur, thus freeing the supervisor to continue with other, more productive activities.

CCV Supervisory Screens

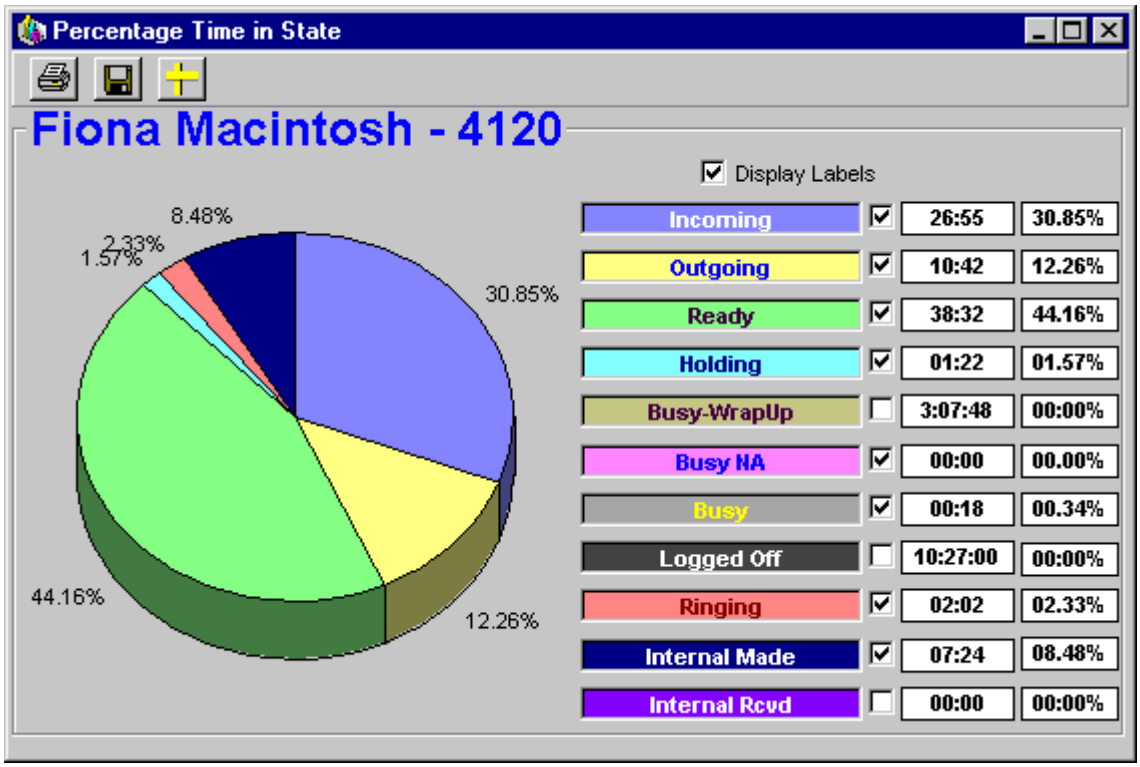
- Alarm Handling.
- BLF Details.
- Extension Activity.
- Callback Request.

Trunk Related Screens

- Trunk Group Monitor.
- Trunk Group Details.
- Real Time Status.
- Group Status (Percentage).
- Individual Trunk Details.

Agent and Queue Based Screens

- Group Monitor
- Agent Group Details
- Real Time Status
- Group Status (Percentage)
- Individual Agent Details
- Percentage Time in State
- Individual Group Details
- Queue Monitor
- Individual DDI/DID Details



Call Center View Real Time Example

© 2008 AVAYA All rights reserved.

CCC Reporter - Historical Reporting

CCC Reporter provides in depth historical reporting on the customer facing department's activity. Report Manager provides standard reports for measuring overall contact center call handling and individual/team performance. Data is retrieved from the database, which provides a source of data limited only by the hard disk space available (SQL only). These standard report templates may be formatted by the user to provide reports daily, weekly, monthly, or any defined time period and by individual, group, or trunk. CCC uses Crystal Reports™ format, which provides ease of use and thin client operation for reporting.

Standard Reports List

- Account Code Log by Agent Group (Graphical)
- Account Code Log by Agent Group
- Account Code Log by DDI (Graphical).
- Account Code Log by DDI.
- Account Code Log by Pilot (Graphical)
- Account Code Log by Pilot.
- Account Code Log by Target (Graphical).
- Account Code Log by Target.
- Agent Activity Trace.
- Agent Activity
- Agent Callback Request.
- Agent Group Busy Status.
- Agent Group Graphical Summary (All Calls).
- Agent Group Graphical Summary.
- Agent Group Member Call Duration Report (All Calls).
- Agent Group Member Duration.
- Agent Group Tabular Summary (All Calls).
- Agent Group Tabular Summary.
- Agent Group Tabular.
- Agent Individual.
- Agent Tabular.
- Customer Tracking by Call Identifier.
- Customer Tracking by CLI.
- DDI Call Duration.
- DDI Distribution by Target.
- DDI Distribution
- DDI Response
- DDI Routing
- DDI Summary.
- External Transferred Account Code.
- Incoming Duration Summary.
- Incoming Pilot Summary.
- Lost Call CLI.
- Outgoing Account Code Costing Log
- Outgoing Account Code Log (Graphical).
- Outgoing Account Code Log.
- Outgoing Most Common Destination by Agent Group.
- Pilot Call Duration.
- Pilot Distribution by Target.
- Pilot Distribution.
- Pilot Response.
- Pilot Routing.
- Pilot Summary (All Calls).
- Pilot Summary
- System Summary.
- Target Graphical Summary.
- Target Member Duration (All Media).
- Target Member Duration.
- Transfer Call Tracking Detail by Agent.
- Trunk Group Activity
- Trunk Group Busy.
- Trunk Group Call Duration.
- Trunk Group Response.
- Trunk Group Summary.
- VM Call Flow Monitor by Call Flow Name.
- VM Call Flow Monitor by Topic.
- VM Call Flow Monitor.
- VM Summary
- Incoming Calls By Target Group
- Plus 3 custom reports.

Report Scheduler

Report Scheduler allows reports to be scheduled to run at a specified date and time, or repeated at regular intervals. Supervisors can schedule reports to be delivered to various places within the contact center. Reports can also be delivered to multiple recipients via email in the following formats; PDF, CSV, XLS, RTF, RPT and Word format. Reports can even be scheduled for delivery to multiple printers within the network at the same time.

© 2008 AVAYA All rights reserved.

Custom Reporting

Custom Reporting allows the business to create reports tailored specifically to the needs of the individual business, providing greater flexibility in the presentation of traffic and agent information. This capability is aimed at the contact center manager who wants to take the statistics to a deeper level in order to make better-informed decisions.

Within Compact Contact Center, custom reporting is available, but requires the purchase of Crystal Reports or Crystal Design software from an authorized Crystal/Business Objects software reseller or distributor. With this software, the designer has the ability to create and load 3 custom reports into the CCC Reporter (no additional license required). Custom reports can be added and subtracted as required. If the business requires greater than 3 custom reports, the following license is required:

- IPO LIC IP 400 CCC DESIGNER RFA LIC:CU

© 2008 AVAYA All rights reserved.

Designing Reports Using Crystal Reports

CCC is designed to work with Crystal Reports™ reporting software package (using Crystal version 9). Crystal Reports is available in four different editions to meet the needs of application developers, IT professionals, and business users. The following is an overview of the types of Crystal products that can be used:

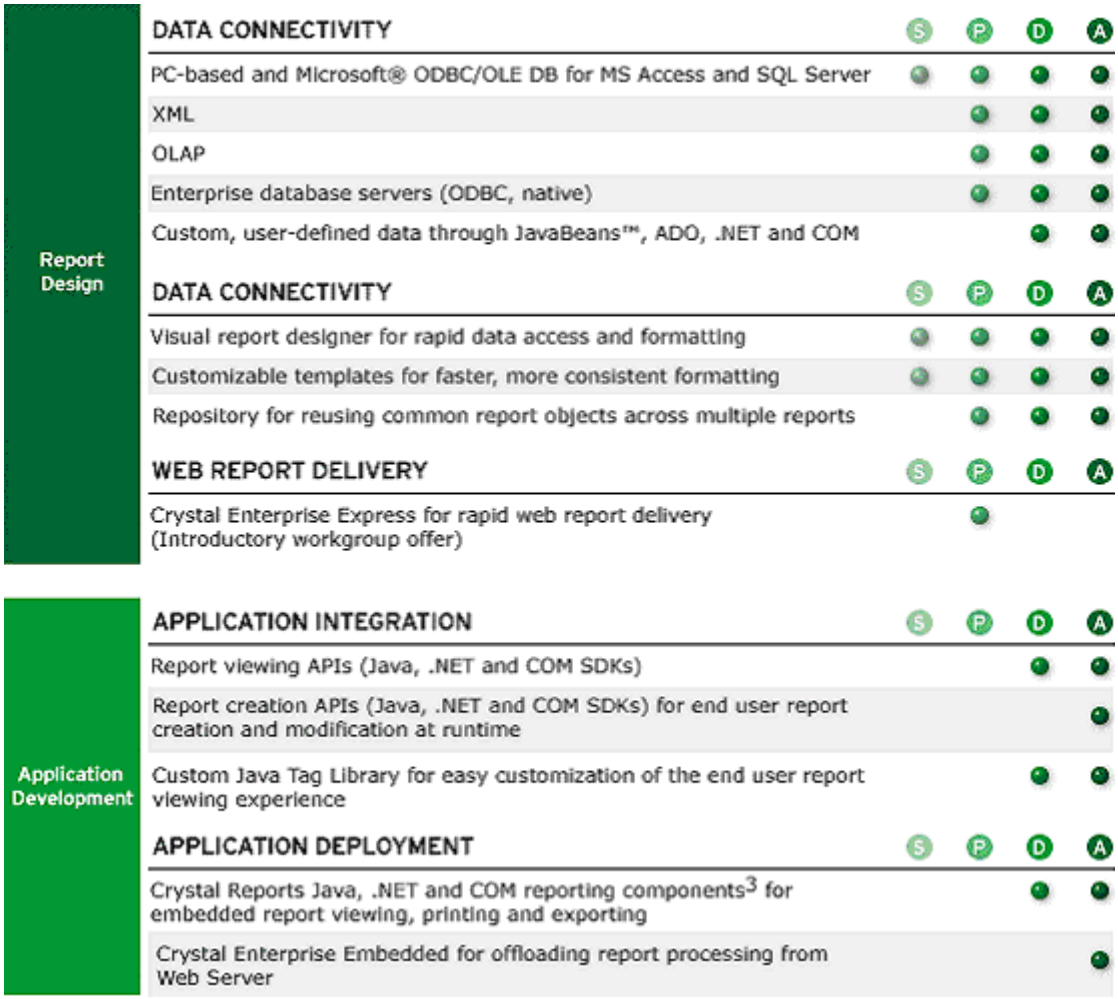
Application Development Solutions

- Advanced Developer – Web development and deployment bundle for integrating and deploying dynamic report creation and viewing capabilities into web applications.
- Developer Edition – For integrating report viewing, printing, and exporting capabilities into applications.

Report Design Solutions

- Professional Edition – For report creation and maintenance based on a large variety of data sources plus out-of-the-box web report delivery for workgroups.
- Standard Edition – For basic report design based on PC-based data sources.

The chart below illustrates some of the key feature differences between the various Crystal Reports 9 editions:


Report Design
DATA CONNECTIVITY
S P D A
PC-based and Microsoft® ODBC/OLE DB for MS Access and SQL Server
XML
OLAP
Enterprise database servers (ODBC, native)
Custom, user-defined data through JavaBeans™, ADO, .NET and COM
DATA CONNECTIVITY
S P D A
Visual report designer for rapid data access and formatting
Customizable templates for faster, more consistent formatting
Repository for reusing common report objects across multiple reports
WEB REPORT DELIVERY
S P D A
Crystal Enterprise Express for rapid web report delivery (Introductory workgroup offer)
Application Development
APPLICATION INTEGRATION
S P D A
Report viewing APIs (Java, .NET and COM SDKs)
Report creation APIs (Java, .NET and COM SDKs) for end user report creation and modification at runtime
Custom Java Tag Library for easy customization of the end user report viewing experience
APPLICATION DEPLOYMENT
S P D A
Crystal Reports Java, .NET and COM reporting components3 for embedded report viewing, printing and exporting
Crystal Enterprise Embedded for offloading report processing from Web Server

For more information on how to purchase Crystal Reports products, go to:

www.businessobjects.com/products/reporting/crystalreports

© 2008 AVAYA All rights reserved.

Crystal Reports Training

Training is available from a number of providers; the following is a sample list.

1. Learning Tree International - www.learningtree.com
2. World-Wide Source for Crystal Training - www.crystal-reports.com
3. Stafford Technology - www.crystaltraining.com

Microsoft CRM™ Reporting Integration New for CCC Version 5

Microsoft CRM™ was introduced in January 2003 and has quickly become the premier CRM application for the Small and Medium Enterprise (SME). Avaya and Microsoft are working together to provide a complete CRM, Communications, and Networking solution for any size of business.

In Compact Contact Center Version 5, in conjunction with the introduction of the IP Office Customer Management solution, Avaya has taken this integration one step further by integrating several Microsoft CRM reports with CCC. Supervisors who operate both systems can now drive any of the 73 CCC reports from the MS-CRM interface, and there are 7 combined reports that utilize both systems data to present a 360° view of the contact center. The 7 MS-CRM reports are listed below:

- Microsoft CRM Sales Reports
  - Opportunity Activity and Notes
  - Contact Activity and Notes
  - Account Activity and Notes
  - Contact Center Summary by State/Province
  - Contact Center Summary by Zip Code/Postal Code
- Microsoft CRM Service Reports
  - Account Activity and Notes
  - Account Service Report

© 2008 AVAYA All rights reserved.

Wallboard Server/Client

Wallboard Manager

Two types of wallboards are available – traditional wall mounted units and PC based wallboards on the agent's PC desktop. Both types of wallboards are managed from Wallboard Manager/Wallboard Server.

Wall Mounted Wallboards are not available in all territories; please check with your Avaya representative for more information.

Additional wallboard clients may be added and distributed across the LAN allowing additional supervisors access to create and schedule wallboard messages.

Traditional Wall Mounted Wallboards

CCC supports two physical wallboards (also known as reader boards or display boards); Spectrum (model 3214C, previously known as the 4120C) and the CCM WB/22. Both wallboards are 22 characters, tri-color, and two-line unit each. Up to 16 wallboards may be driven from the wallboard server. The Spectrum wallboard, when purchased as a Master Kit, will provide a communications module for use with the boards which are connected serialy. For those using the Wallboard/22, the communications card is shipped with a single cable able to drive the wallboards.

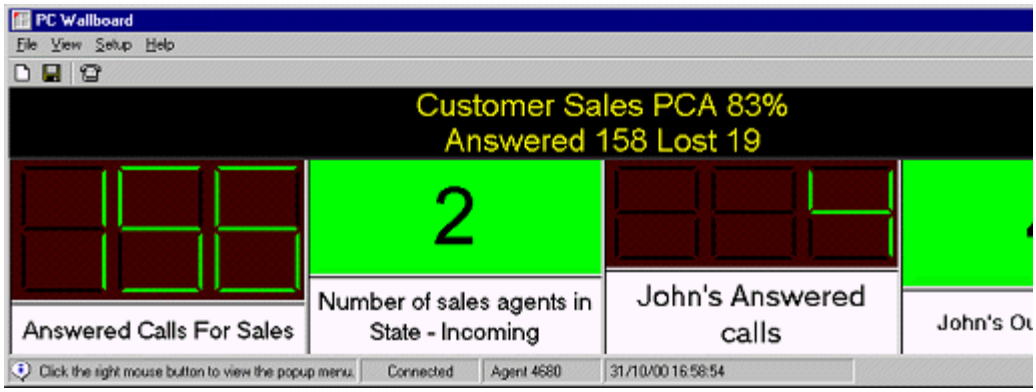
In addition to the physical Spectrum wall-mounted wallboard an IP Office license is required when being used with CCC. This IP Office license supports 4 x Spectrum wall-mounted wallboards. If more than 4 wall-mounted wallboards are required additional license keys must be purchased (each license key supports 4 wallboards at a time). A maximum of 16 wall-mounted wallboards can be supported.

Description	Short code	Material code
Wallboard/22	IND DISP CCM WALLBRD 22 GB	700040173
Wallboard Manager Communications card	IND CP CCM WALLBRD	700038854
IP 400 CCC Wallboard 4 RFA License key required supporting 4 wallboards.	IPO LIC IP400 CCC WALLBRD 4 RFA LIC: CU	176196

PC Wallboard

The PC Wallboard delivers wallboard functionality to the contact center manager and contact center agent's desktop, but with the benefit of each agent being able to configure and monitor a personalized view of the contact center via their own PC wallboard. Supervisors can provide one template for all users in order to standardize the view that agents obtain when starting PC Wallboard.

A CCC agent is able to split their PC Wallboard into twenty (20) different variables that allow different measures of groups and agents in real-time. The data that is presented is identical to that of the physical wallboard. Examples of this are Answered Calls, Longest Call Waiting, Agents logged in, and Lost Calls.



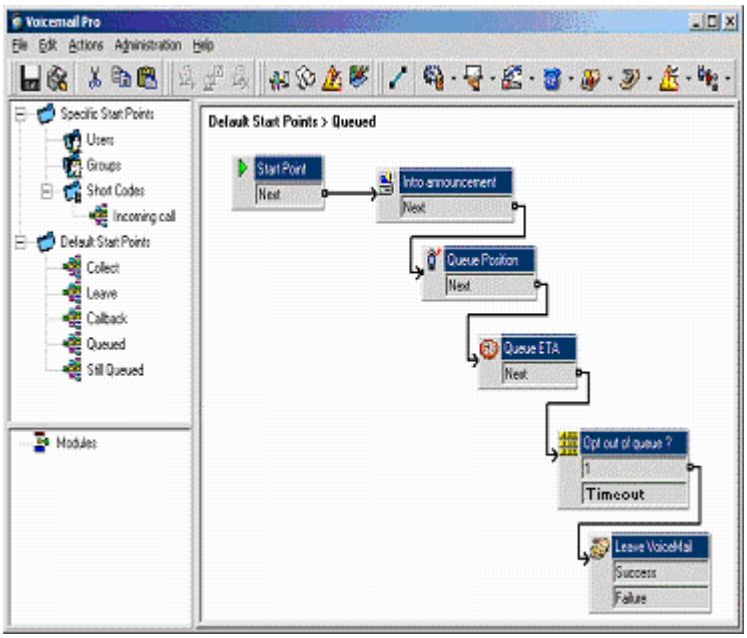
PC Wallboard Example

© 2008 AVAYA All rights reserved.

Queuing Announcements

VoiceMail Pro provides system wide messages and announcements programmed by VoiceMail Pro call flows. Through call flows it is possible tailor the pre-connection call experience that a customer receives when calling in. By using the functionality provided by Voicemail Pro's call in-queue announcements, supervisors may create sophisticated queue and call routing plans with access to a host of features such as message taking, interview services, and the ability to play estimated time to answer or queue position information to customers.

The VoiceMail Pro application provides Queue Handling facilities, allowing incoming Hunt Group calls to be answered when department, group or individual telephones are busy. Customers entering a queue are played a message informing them of the situation and then hear hold music (internally generated or from an external source), while being regularly updated. Two unique messages may be recorded for each Hunt Group (queue entry and queue update message). Queue announcements can also provide time in system, time in queue, position in queue and estimated time to answer to the caller. It always gives the caller the option to opt out of the queue and leave a message at any time if desired.



© 2008 AVAYA All rights reserved.

Chapter 14.
CRM Integration

© 2008 AVAYA All rights reserved.

14. CRM Integration

Introduction

Avaya and Microsoft enjoy a global partnership. Avaya's innovative voice communications and applications based on Microsoft Windows .NET and Dynamics CRM platform, are enabling small medium business to become more effective and profitable. As a Gold Certified Partner and thought leader, Avaya, in partnership with Microsoft, continues to deliver a broad spectrum of technologies that are reliable, scalable and secure.

Avaya – Microsoft Dynamics® CRM 3.0 Integration

The Avaya Microsoft™ CRM Integration Solution allows a business to connect Microsoft Dynamics® CRM 3.0 to Avaya IP Office. It integrates contact points in such a way that will transform the way your business interacts with your customers, this is accomplished by integrating incoming calls directly to the desktop of the user through the use of screen pop technology and by providing outbound dial capability directly from the Microsoft CRM entity.

The Avaya Microsoft CRM Integration Solution requires the following applications to be installed on the Server PC prior to installing the Avaya Microsoft CRM Integration Solution.

- Microsoft Dynamics® CRM 3.0
- IIS 5.0 +

The Avaya Microsoft CRM Integration Solution requires the following applications installed on the Client PC. The client machine will be checked at installation for these components and they will be installed if not found.

- Microsoft .NET 2.0
- IP Office TAPI 2.1 Driver (1.0.0.27)

The Avaya Microsoft CRM Integration Solution is supported on the following client operating systems:

- Microsoft Windows 2000™ Professional
- Microsoft Windows XP™ Professional

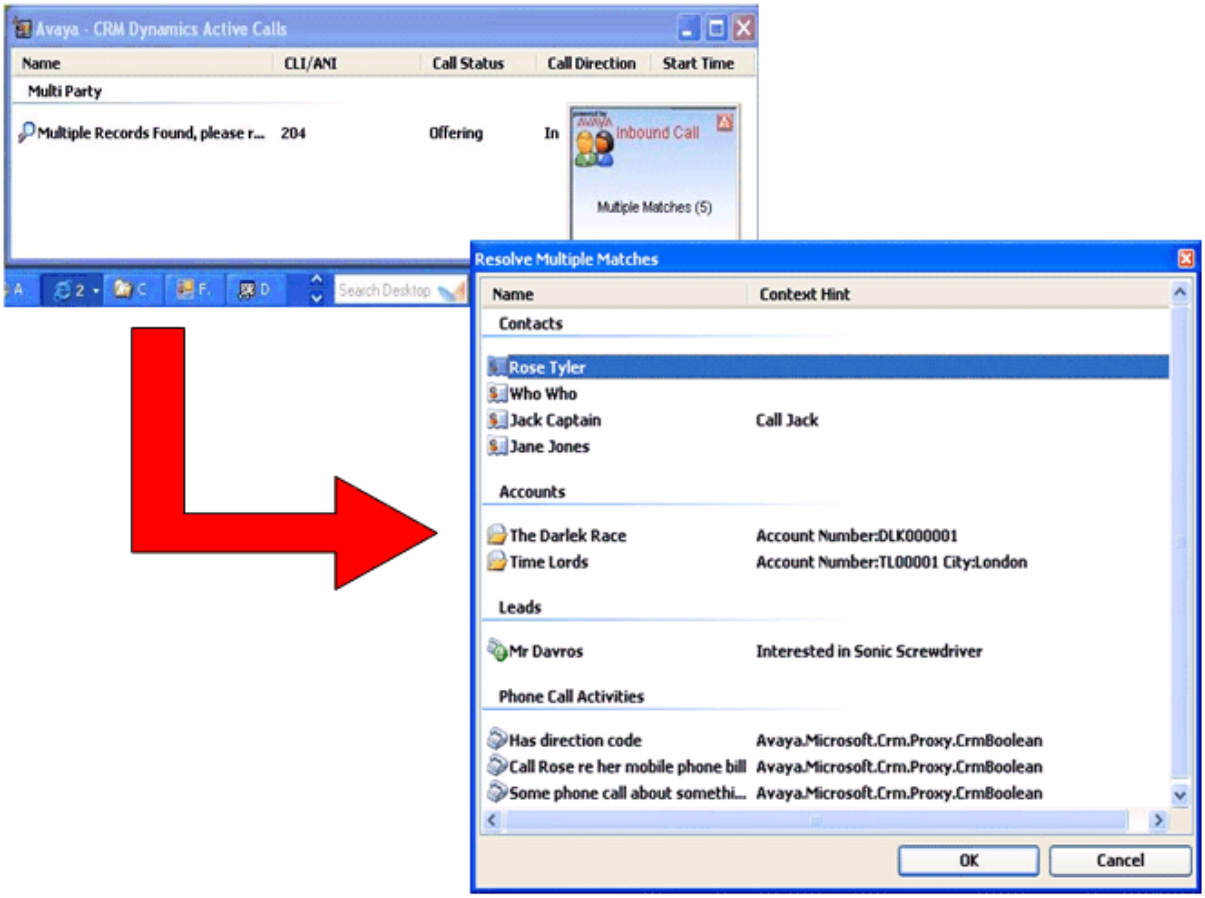
© 2008 AVAYA All rights reserved.

Inbound Call Operation

A user can set up their integration to provide inbound screen pops for the following screens within Microsoft CRM™:

- **Contacts**
- **Accounts**
- **Leads**
- **Phone Call Activities**

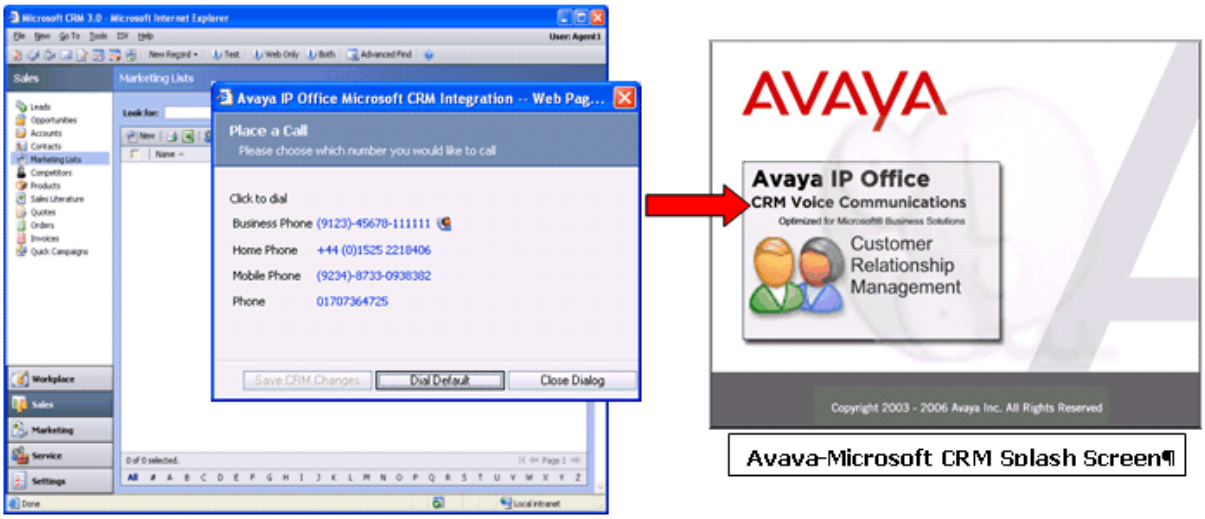
The user can define what actions to take when an inbound call matches multiple screens, this is accomplished through the use of an Answer Bar that allows the user to select which screen to "pop" into, as identified below:



© 2008 AVAYA All rights reserved.

Outbound Call Operation

Outbound calls are tightly integrated with the Microsoft CRM screen for quick, easy dialing directly from the application.


Microsoft CRM 3.0 - Microsoft Internet Explorer
Avaya IP Office Microsoft CRM Integration -- Web Pag...
Place a Call
Please choose which number you would like to call
Click to dial
Business Phone (9123)-45678-111111
Home Phone +44 (0)1525 2218406
Mobile Phone (9234)-8733-0938382
Phone 01707364725
Save CRM Changes
Dial Default
Close Dialog
AVAYA
Avaya IP Office
CRM Voice Communications
Customer Relationship Management
Copyright 2003 - 2006 Avaya Inc. All Rights Reserved

Avava-Microsoft CRM Splash Screen¶

Customer Benefits

- Link customer information with the touch points used to interact with them
- Handling calls more effectively—reducing and eliminating long hold times, multiple transfers, abandoned calls
- Support employees across the business—everyone working off the same customer information
- Getting calls to the right person at the right time with the right information
- Remembering every customer interaction

© 2008 AVAYA All rights reserved.

© 2008 AVAYA All rights reserved.

Chapter 15. Management Utilities

© 2008 AVAYA All rights reserved.

15. Management Utilities

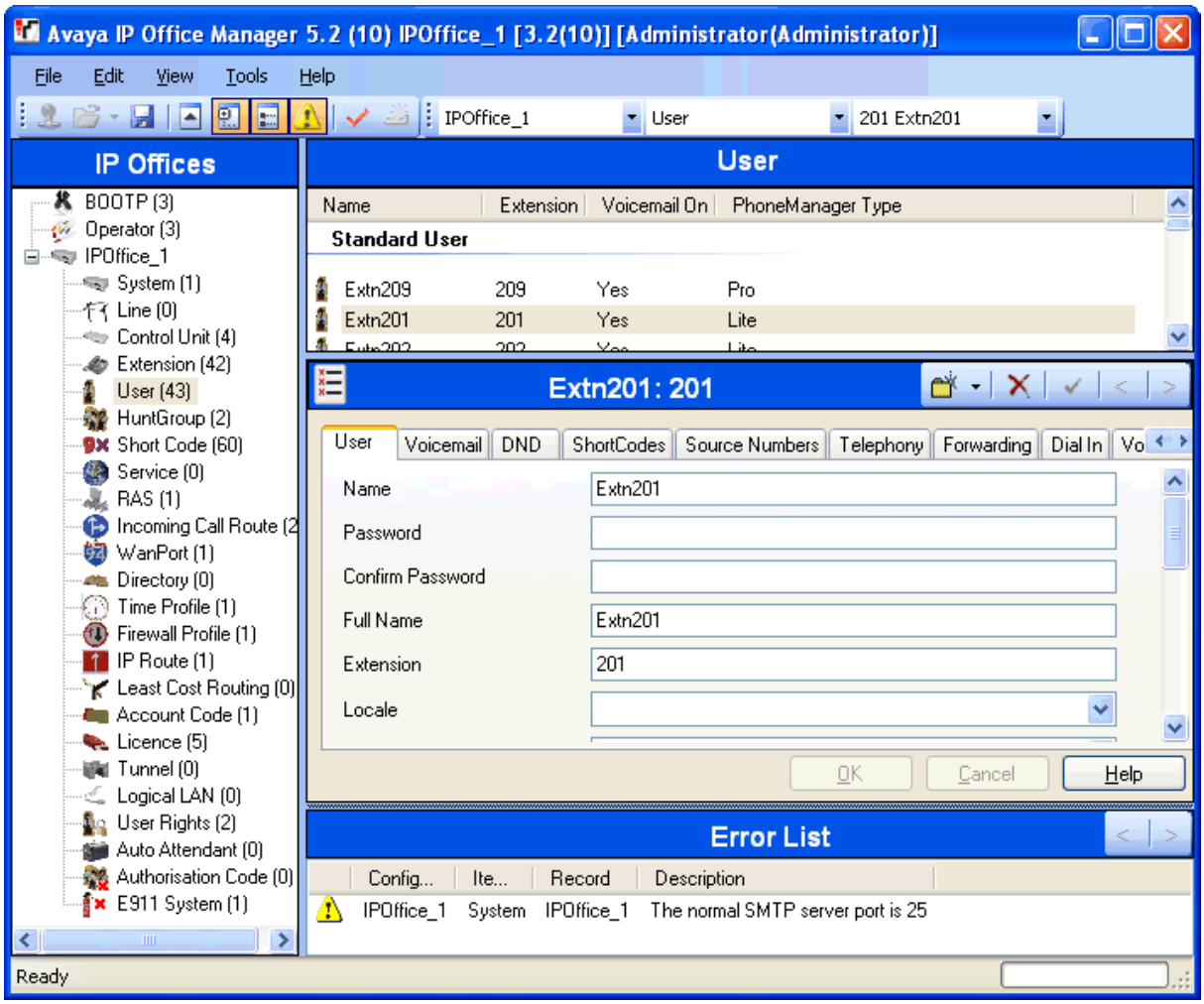
This section gives an overview of the management applications that are common to all IP Office platforms.

- **IP Office Manager** 266
  IP Office's main configuration tool.
- **Monitor** 269
  A trace utility for trouble shooting.
- **SNMP** 270
  Alerts and alarms from IP Office systems to SNMP tools or to SMTP email.
- **CDR** 271
  Outputs call detail records direct to an attached printer or separate PC.
- **IP Office SMDR** 272
  Outputs station message detail records (SMDR) for off switch processing.
- **System Status Application (SSA)** 273
  Outputs call detail records for off A diagnostic tool to monitor and check the status of IP Office systems.

© 2008 AVAYA All rights reserved.

IP Office Manager

This application is IP Office's main configuration tool. Using a Windows Graphical User Interface, Manager provides an intuitive interface for installation, configuration and subsequent moves and changes. As with all IP Office applications, the Manager is multi-lingual and coupled with the ability to use the application both locally and remotely, it is possible for an administrator to manage any of their IP Offices from any country using their local language preference. Access to each IP Office is protected by passwords and definable user rights. This allows Manager to operate according to the individual administrator's level of expertise.

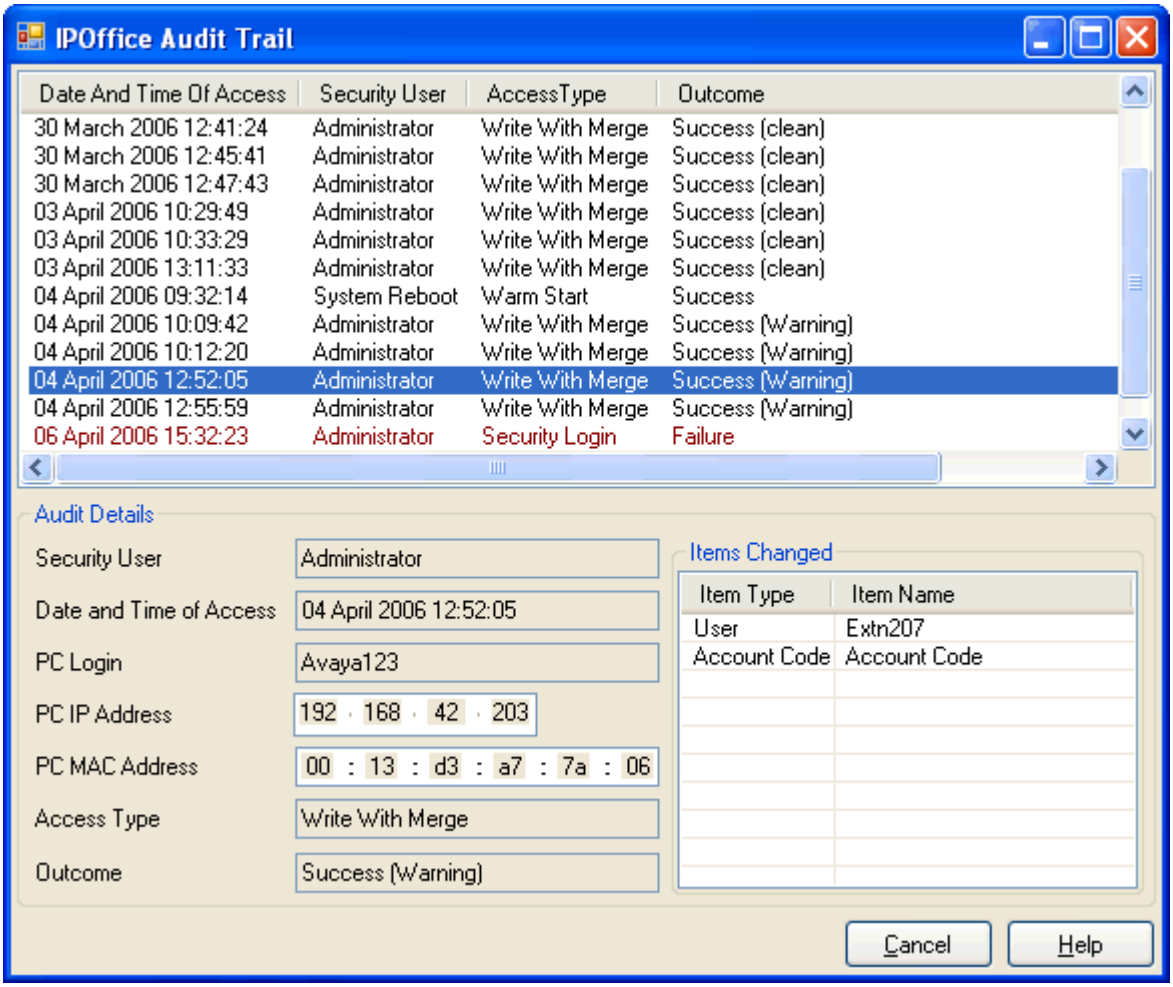


The IP Office Manager operates on a local copy of the IP Office configuration file. Configurations are prepared and reviewed 'off line' before committing to the IP Office. This has the benefit of ensuring a backup copy of the system configuration is always available for disaster recovery.

© 2008 AVAYA All rights reserved.

IP Office has a built-in audit trail that tracks changes to the system configuration, and who has made them. Manager can display the audit trail to assist with problem resolution. The Audit trail records the last 15 changes in the configuration and records the following elements:

- Configuration Changed - For configuration changes, the log will report at a high level on all configuration categories (users, hunt group...) that have been changed.
- Configuration Erased
- Configuration merged
- Reboot – user instigated reboot.
- Upgrade
- Cold Start
- Warm Start
- Write at HH:MM – This is when the administrator saved the configuration via the schedule option
- Write with Immediate Reboot
- Write with Reboot When Free


IPOffice Audit Trail
Date And Time Of Access
Security User
AccessType
Outcome
30 March 2006 12:41:24 Administrator Write With Merge Success (clean)
30 March 2006 12:45:41 Administrator Write With Merge Success (clean)
30 March 2006 12:47:43 Administrator Write With Merge Success (clean)
03 April 2006 10:29:49 Administrator Write With Merge Success (clean)
03 April 2006 10:33:29 Administrator Write With Merge Success (clean)
03 April 2006 13:11:33 Administrator Write With Merge Success (clean)
04 April 2006 09:32:14 System Reboot Warm Start Success
04 April 2006 10:09:42 Administrator Write With Merge Success (Warning)
04 April 2006 10:12:20 Administrator Write With Merge Success (Warning)
04 April 2006 12:52:05 Administrator Write With Merge Success (Warning)
04 April 2006 12:55:59 Administrator Write With Merge Success (Warning)
06 April 2006 15:32:23 Administrator Security Login Failure
Audit Details
Security User
Administrator
Date and Time of Access
04 April 2006 12:52:05
PC Login
Avaya123
PC IP Address
192 . 168 . 42 . 203
PC MAC Address
00 : 13 : d3 : a7 : 7a : 06
Access Type
Write With Merge
Outcome
Success (Warning)
Items Changed
Item Type
Item Name
User
Extn207
Account Code
Account Code
Cancel
Help

© 2008 AVAYA All rights reserved.

Manager is also used for maintenance functions such as:

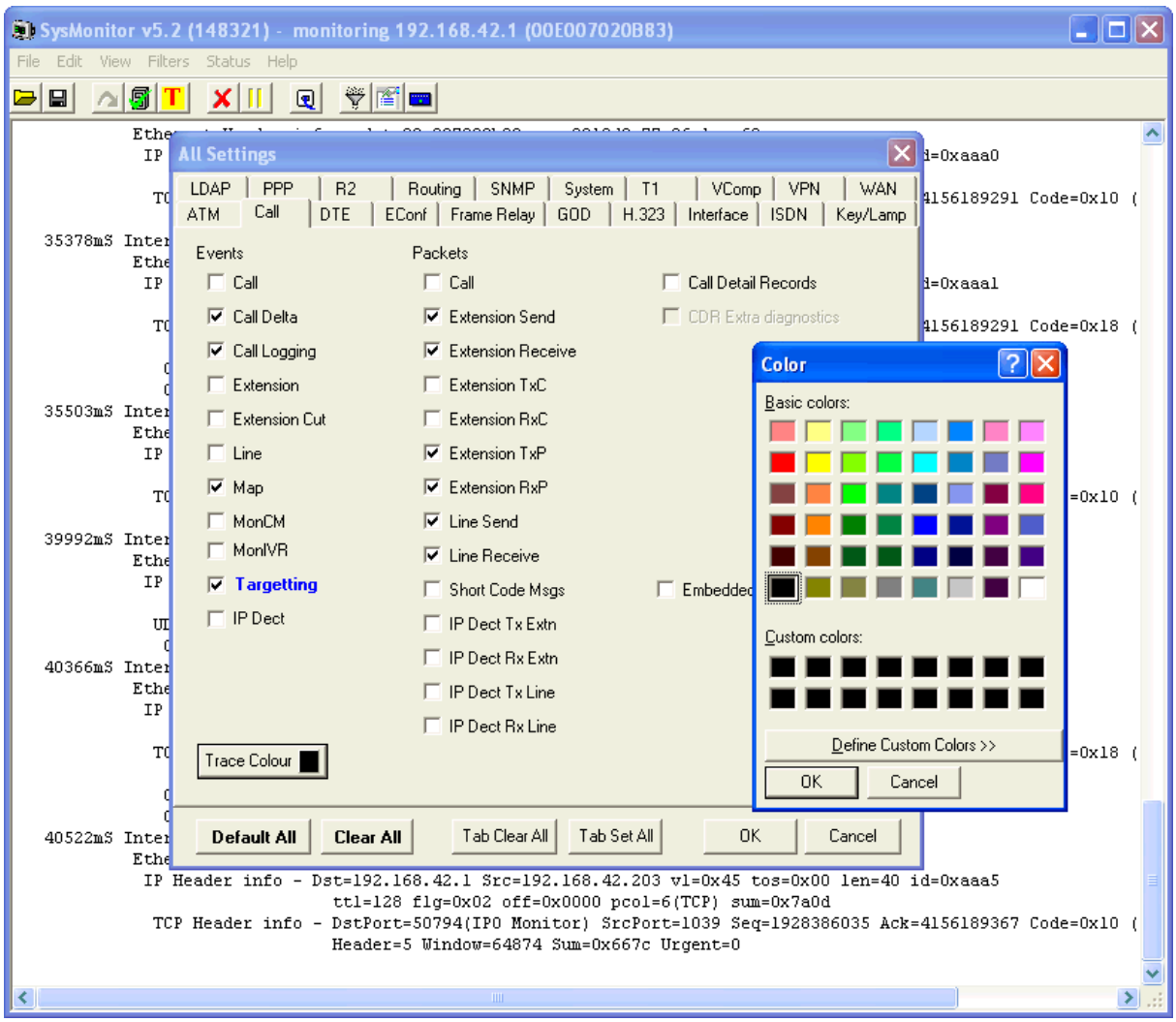
- Upgrade to the IP Office system software.
  - Systems running 2.1 or later have the added benefit of being able to send software over an IP network link to a system and have it validated before committing to the upgrade
- IP Office Manager 3.2 is backwards compatible with systems from release 2.1 onwards to allow a single management application.
- Importing and Exporting IP Office configuration information in ACSII-CSV files. Manager will create files for the following data
  - Configuration.csv which is a complete list of items as per Manager 5.1 and earlier
  - Directory.csv containing fields NAME, NUMBER
  - HuntGroup.csv containing fields HUNT GROUP NAME, HUNT GROUP EXTENSION, GROUP, HUNT, ROTARY, IDLE, QUEUING, VOICEMAIL, BROADCAST MESSAGES, EMAIL ADDRESS
  - License.csv is import only containing fields LICENCE OPTION, LICENCE KEY
  - ShortCode.csv containing fields SHORT CODE, TELEPHONE NUMBER, FEATURE NAME
  - User.csv containing fields NAME, EXTENSION NUMBER, USER RIGHT, EMAIL ADDRESS
- User templates for rapid programming and user rights for setting up user access levels

© 2008 AVAYA All rights reserved.

Monitor

The IP Office Monitor application is a real-time maintenance utility to assist with IP Office trouble-shooting. As the application connects to the IP Office over an IP connection it can be used from both local (LAN) and remote locations (WAN).

A simple interface allows an engineer to select which protocols and interfaces are to be monitored and decoded. The trace can either be captured directly to screen or as a log file for later analysis. Traces from different protocols can be color coded to improve the clarity of large log files. In addition to monitoring, the application captures system alarms and will display an activity log of the last 20 alarms that have occurred.

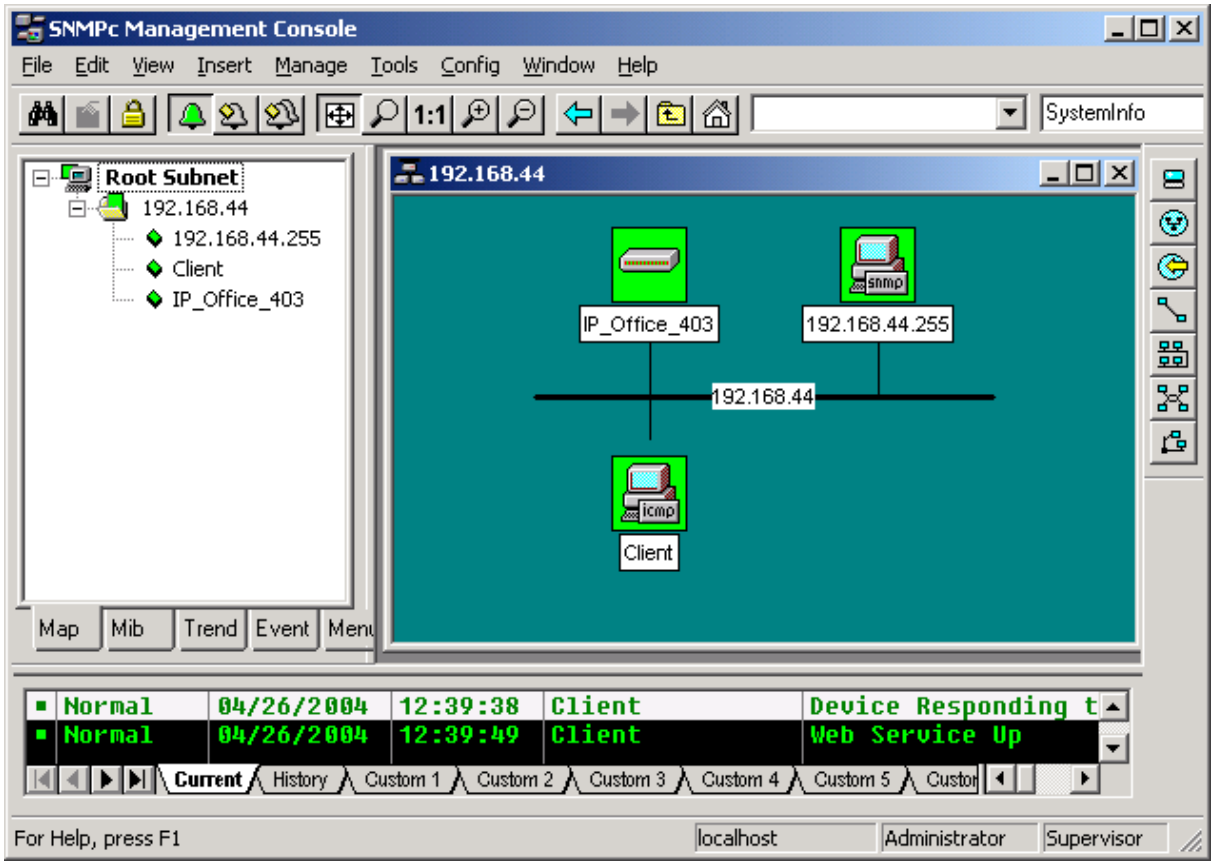


© 2008 AVAYA All rights reserved.

Simple Network Management Protocol (SNMP)

SNMP is an industry standard designed to allow the management of data equipment from different vendors using a single Network Manager application. The Network Manager will periodically poll equipment to solicit a response, if no response is received an alarm is raised. In addition to responding to polls, IP Office monitors the state of its Extensions, Trunk cards, Expansion Modules (except WAN3 module) and Media cards so that if an error is detected IP Office will notify the Network Manager. IP Office allows two separate Network Managers to be configured so that both a customers Network Manager and a Maintainers Network Manager to be notified of the same alarm condition. As the IP Office solution comprises many applications, the core software notifies SNMP events from both VoiceMail Pro and Embedded voicemail to warn of approaching storage capacity limits.

IP Office has been tested against CastleRock's SNMPc-EE™ and HP's Network Node Manager (part of the OpenView application suite). Avaya's 'Integrated Management Suite' also uses HP's Network Node Manager.



On customer sites where SNMP management is not available, IP Office can email events using up to 3 email addresses each containing a different set of alarms. The following system event categories can be chosen for email notification, if installed on the system:

- **Generic**
- **Trunk lines**
- **Embedded Messaging Card**
- **VCM**
- **Expansion modules**
- **Applications**
- **License**
- **Phone change**
- **CSU Loop-Back**

IP Office sends email notifications directly to the email server; no additional PC client is needed.

© 2008 AVAYA All rights reserved.

CDR

For IP Office customers that want to capture simple call details, the system can output Call Detail Records (CDR) to a designated IP address and port. The records that can be included by IP Office CDR output are listed below:

- **Date Records**
  A date record is sent each time a CDR connection is started and then once a day (at midnight). The date can be in month/day or day/month format, as selected on the System | CDR/SMDR tab.
- **Call Detail Records**
  Call detail records are sent at the termination of a call (in 5 second increments). For some formats, additional fields can be selected using the Normal, Enhanced, or ISDN options on the System | CDR/ SMDR tab.

Depending upon the selected report format and options, there are a number of different fields available within the CDR, they are listed as follows (please review the IP Office Manager documentation for further information):

- Access Code Dialed
- Access Code Used
- Account Code
- BCC (Bearer Capability Class)
- Calling Number
- Calling Number/Incoming Trunk Access Code
- Carriage Return
- Condition Code
- Dialed Number
- Duration
- Feature Flag
- Incoming Circuit ID
- Incoming Trunk Access Code
- Line Feed
- Null
- Outgoing Circuit ID
- Space
- Time

© 2008 AVAYA All rights reserved.

SMDR

For more formal call logging and reporting, the IP Office Station Message Detail Report (SMDR)output is used by third party applications for many call accounting applications. IP Office SMDR provides much greater details of the call, including duration, ring time, hold time, and transfer information.

From Release 4.2 onwards, IP Office can output SMDR events directly as well as through a separate Windows service included in the IP Office Delta Server application. To generate SMDR events directly, choose SMDR from the System | CDR/SMDR tab in Manager.

The IP Office Delta Server (SMDR) application is provided on the Admin portion of the IP Office CD/DVD set. It allows the detail of all calls to be sent to a file on the PC. Both methods allow the details of calls to be sent over an IP network to a TCP/IP port.

Third party applications use this data to allocate costs to departments, analyze trunk capacity, report usage against account codes etc. One IP Office SMDR (Delta Server) is required for each site requiring the use of call accounting software. Please refer to the Technical Specifications section for the Delta Server requirements.

SMDR Diagnostics

Time Of Call Arrival	Call Duration	Ring Time	CLI	Dir.	DDI	DDI	Account Code	Internal	Call ID	More	P1 ID	P1 Name	P2 ID	P2 Name	Hold Time	Park Time
2004/10/19 07:47:07	00:00:00	0	211	O	215	215		1	6	0	E215	Extn215	E215	Extn215	0	0
2004/10/19 07:47:07	00:00:00	0		O				1	100	0	E-1	No Name			0	0
2004/10/19 07:46:56	00:00:10	0	215	I	215	215		0	6	0	V9551	CO Channel 1	E215	Extn215	0	0
2004/10/19 07:46:54	00:00:09	1	211	I	369	369		0	7	0	V9551	CO Channel 1	E211	Extn211	0	0
2004/10/19 07:46:56	00:00:07	0	211	I	9551	9551		0	7	0	V9551	CO Channel 1	E369	Extn369	0	0

Sample IP Office SMDR Information Output

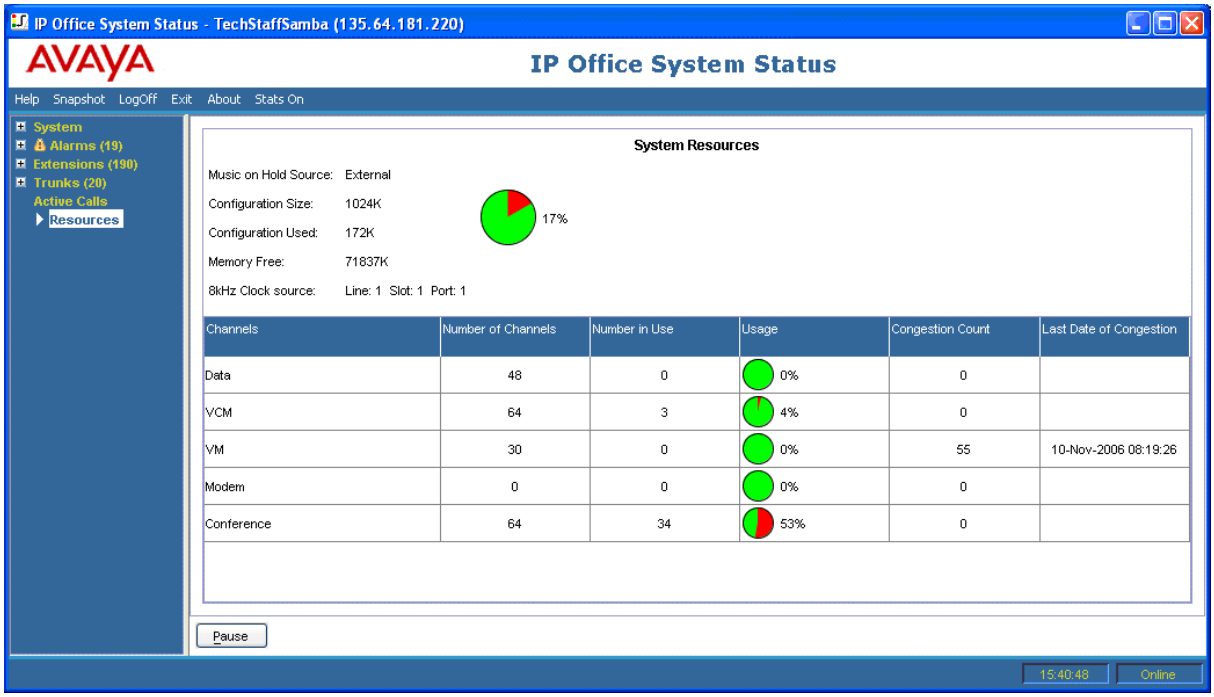
© 2008 AVAYA All rights reserved.

System Status Application

The System Status Application (SSA) is a diagnostic tool for system managers and administrators to monitor and check the status of IP Office systems locally or remotely. SSA shows both the current state of an IP Office system and details of any problems that have occurred. The information reported is a combination of real-time events, historical events, status and configuration data to assist fault finding and diagnosis. SSA provides real-time status, historic utilization and alarm information for ports, modules and expansion cards on the system. SSA connects to all variants of IP Office running release 4.0, using an IP connection that can be remote or local. Modem connections at 14.4kbps or above are supported for remote diagnostics.

SSA provides information on the following:

- **Alarms**
  SSA displays all alarms which are recorded within IP Office for each device in error. The number, date and time of the occurrence is recorded. The last 50 alarms are stored within IP Office to avoid need for local PC.
- **Call Details**
  Information on incoming and outgoing calls, including call length, call ID and routing information.
- **Extensions**
  SSA details all extensions (including device type and port location) on the IP Office system. Information on the current status of a device is also displayed.
- **Trunks**
  IP Office trunks and connections (VoIP, analog and digital) and their current status are displayed. For VoIP trunks, QoS information is also displayed (e.g. round trip delay, jitter and packet loss)
- **System Resources**
  IP Office includes central resources that are utilized to perform various functions. Diagnosing these resources is often critical to the successful operation of the system. This includes details on resources for VCM, Voicemail and conferencing.


IP Office System Status - TechStaffSamba (135.64.181.220)
AVAYA
IP Office System Status
Help Snapshot LogOff Exit About Stats On
System
Alarms (19)
Extensions (190)
Trunks (20)
Active Calls
Resources
System Resources
Music on Hold Source: External
Configuration Size: 1024K
Configuration Used: 172K
Memory Free: 71837K
8kHz Clock source: Line: 1 Slot: 1 Port: 1
17%
Channels | Number of Channels | Number in Use | Usage | Congestion Count | Last Date of Congestion
Data | 48 | 0 | 0% | 0 |
VCM | 64 | 3 | 4% | 0 |
VM | 30 | 0 | 0% | 55 | 10-Nov-2006 08:19:26
Modem | 0 | 0 | 0% | 0 |
Conference | 64 | 34 | 53% | 0 |
Pause
15:40:48
Online

SSA can be launched independently or from IP Office Manager and there can be up to two (2) SSA clients connected to an IP Office unit at one time.

Note: SSA is not a configuration tool for IP Office systems. For information on configuration, refer to IP Office Manager.

In IP Office 4.2, System Status Application (SSA) has a new real-time configuration alarms screen which shows the following new alarms:

- Duplicate Numbers in Small Community Networking dial plan.
- "No routing for Line" - Incoming Line Group has no associated Incoming Call Route.
- Call Routed to a non-existent Start Point in VoiceMail Pro.
    - SSA displays the following new resource availability alarms in IP Office 4.2:

© 2008 AVAYA All rights reserved.

    - Attempt to use a feature for which no license is installed.
- Voicemail storage full or nearly full.
- The following new screens are provided in SSA in IP Office 4.2:
    - Control Unit Audit - details the last 15 accesses from Manager.
    - Feature Key and License Status.
    - Directory, including Users on the SCN.
    - Real time Voicemail port status.
    - Voicemail Summary/Detail information.
    - Real time Mailbox status.
    - IP Route and Service status.
    - Tunnel Status.

© 2008 AVAYA All rights reserved.

© 2008 AVAYA All rights reserved.

Chapter 16. Configurations

© 2008 AVAYA All rights reserved.

16. Configurations

IP500 Control Unit (700417207)

Includes: 4 x front slots for combinations of extension/VCM cards and trunk daughter cards, 1 x smart card slot for locale settings and license feature key, 1 x compact flash slot for embedded voicemail option, 2-port Layer-3 LAN switch, 9-pin DTE serial port for system diagnostics, 3.5 mm jack for Music-on-Hold audio input and 2-switch external door-relay control port. 8 x external expansion module ports to support additional analog trunks, digital or analog extensions. Includes auto ranging internal power supply. Regional power cord and software/documentation CD pack not included. Only one variant of control unit is available, but regional locale is determined by the appropriate smart card feature key (mandatory):

- **IP Office 500 Software License Feature Key Mu-Law *(700417470)***
  Configures the control unit for Mu-law voice encoding and US locale settings.
- **IP Office 500 Software License Feature Key A-Law *(700417488)***
  Configures the control unit for A-law voice encoding and multi-country locale settings.

IP412 Control Units

Includes: 2-port Layer-2 LAN switch, 9-pin DTE serial port for license feature key and system diagnostics, 37-pin WAN port, 3.5 mm jack for Music-on-Hold audio input and 2-switch external door-relay control port. Internal expansion slots to support 1 x 12-port remote access modem module and 2 x Voice Compression Modules (including VCM24 and 30 for non-blocking IP/dual-PRI applications). 12 x external expansion module ports to support additional analog trunks, WAN interfaces, digital or analog extensions. Includes 60W earthed external power supply. Regional power cord and software/documentation CD pack not included.

- **IP412 Office Mu-Law Base Unit *(700350408)***
  Mu-law voice encoding base unit pre-configured for US locale settings. 2 x trunk module slots to support US T1 PRI and 4-port analog trunk cards.
- **IP412 Office A-Law Base Unit *(700234479)***
  A-law voice encoding base unit pre-configured for multi-country locale settings. 2 x trunk module slots to support Euro-ISDN BRI, E1/PRI and 4-port analog trunk cards.

© 2008 AVAYA All rights reserved.

IP Office External Expansion Modules

Except where noted, all the following are supported by the IP406 V2, IP412 and IP500 control units. Note that external expansion modules are only supported by the IP500 when running in IP Office Professional Edition mode.

IP500

- **IP Office 500 Expansion Module Phone 16** ***(700449507)***
  Adds an additional 16 analog Plain Ordinary Telephone ports to control units.
- **IP Office 500 Expansion Module Digital Station 16** ***(700449499)***
  Adds an additional 16 Digital Station ports to control units.
- **IP Office 500 Expansion Module Phone 30** ***(700426224)***
  Adds an additional 30 analog Plain Ordinary Telephone ports to control units.
- **IP Office 500 Expansion Module Digital Station 30** ***(700426216)***
  Adds an additional 30 Digital Station ports to control units.
- **IP500 BRI So8 Module** ***(700449515)***
  Provides 8 ISDN BRI S-interface device lines to the desktop.
- **IP500 Analog Trunk 16 -** ***North America only (700449473)***
  Provides an additional 16 Analog trunks (loop start or ground start) and two power fail sockets.

IP400

- **IP400 Phone 8 Module V2** ***(700359896)***
  Adds an additional 8 analog Plain Ordinary Telephone ports to control units.
- **IP400 Phone 16 Module V2** ***(700359904)***
  Adds an additional 16 analog Plain Ordinary Telephone ports to control units.
- **IP400 Phone 30 Module V2** ***(700359912)***
  Adds an additional 30 analog Plain Ordinary Telephone ports to control units.
- **IP400 Digital Station 16 Module V2** ***(700359839)***
  Adds an additional 16 Digital Station ports to control units.
- **IP400 Digital Station 30 Module V2** ***(700359847)***
  Adds an additional 30 Digital Station ports to control units.
- **IP400 Analog Trunk 16 - North America only** ***(700211360)***
  Provides an additional 16 Analog trunks (loop start or ground start) and two power fail sockets.
- **IP400 Analog Trunk 16 EU** ***(700241680)***
  Provides an additional 16 Analog trunks (loop start) and two power fail sockets. European CTR21 specification.

© 2008 AVAYA All rights reserved.

IP500 Voice Compression Modules

- **IP Office 500 Media Card Voice Coding Module 32 *(700417389)***
  Voice Compression Module required for IP trunks and extensions. 4 channels are enabled by default. Additional channels up to the maximum of 32 are enabled through license keys. Includes 128ms echo cancellation.
- **IP Office 500 Media Card Voice Coding Module 64 *(700417397)***
  Voice Compression Module required for IP trunks and extensions. 4 channels are enabled by default. Additional channels up to the maximum of 64 are enabled through license keys. Includes 128ms echo cancellation.

IP500 Base Cards

Only supported in the IP500.

- **IP Office 500 Extension Card Digital Station 8 *(700417330)***
- **IP Office 500 Extension Card Phone 8 *(700417231)***
- **IP Office 500 Extension Card Phone 2 *(700431778)***

- **IP500 Media Card Voice Compression Module 32 *(700417389)***
  Voice Compression Module required for IP trunks and extensions. 4 channels are enabled by default. Additional channels up to the maximum of 32 are enabled through license keys. Includes 128ms echo cancellation.
- **IP500 Media Card Voice Compression Module 64 *(700417397)***
  Voice Compression Module required for IP trunks and extensions. 4 channels are enabled by default. Additional channels up to the maximum of 64 are enabled through license keys. Includes 128ms echo cancellation.
- **IP Office 500 Carrier Card *(700417215)***
  Supports a range of IP400 trunk cards.

IP500 Trunk Daughter Cards

Only supported in the IP500. Each requires an IP500 base card other than the IP500 Carrier card.

- **IP Office 500 Trunk Card Analog 4 Universal *(700417405)***
- **IP Office 500 Trunk Card Primary Rate 1 Universal *(700417439)***
- **IP Office 500 Trunk Card Primary Rate 2 Universal *(700417462)***
- **IP Office 500 Trunk Card Basic Rate 4 Universal *(700417413)***
- **IP Office 500 Trunk Card Basic Rate 8 Universal *(700417421)***

IP400 Voice Compression Modules

All of the following can be installed in the IP Office 500 using the IP500 Legacy Card Carrier *(700417215)*.

- **Voice Compression Module 4 *(700359854)***
  4 Channel Voice Compression module required for IP trunks and extensions. Includes 64ms echo cancellation.
- **Voice Compression Module 8 *(700359862)***
  8 Channel Voice Compression module required for IP trunks and extensions. Includes 64ms echo cancellation.
- **Voice Compression Module 16 *(700359870)***
  16 Channel Voice Compression module required for IP trunks and extensions. Includes 64ms echo cancellation.
- **Voice Compression Module 24 *(700359888)***
  24 Channel Voice Compression module required for IP trunks and extensions. Includes 64ms echo cancellation.
- **Voice Compression Module 30 *(700293939)***
  30 Channel Voice Compression module required for IP trunks and extensions. Includes 25ms echo cancellation.

IP400 Modems cards

- **IP400 Modem 12 (700343452)**
  Internally fitted card allowing twelve simultaneous V.90 modem calls. Not supported on the IP500.

© 2008 AVAYA All rights reserved.

IP400 Trunk Interface Cards

Except where noted, all of the following can be installed in the IP500 using the IP500 Legacy Card Carrier *(700417215)*.

- **IP400 BRI-8 (UNI) (700262017)**
  Interface card for the Small Office Edition, IP406 and IP412 providing 4 x ISDN T-Bus Basic Rate Interface ports (8 lines).
- **IP400 PRI 30 E1 (1.4) (700272461)**
  Interface card for the IP406 and IP412 providing 1 x ISDN Primary rate port (30 lines).
- **IP400 PRI 30 E1R2 RJ45 - CALA (700241631)**
  Interface card for the IP406 and IP412 providing 1 x E1R2 Primary rate port (30 lines). RJ45 termination.
- **IP400 PRI 30 E1R2 COAX - CALA (700241656)**
  Interface card for the IP406 and IP412 providing 1 x E1R2 Primary rate port (30 lines). Co-Ax termination. Not supported on the IP500.
- **IP400 Dual PRI E1 (700185184)**
  Interface card for the IP406 and IP412 providing 2 x ISDN Primary rate ports (60 lines).
- **IP400 PRI T1 (700185200)**
  Interface card for the IP406 and IP412 providing 1 x T1/PRI port (24 lines).
- **IP400 Dual PRI T1 (700185218)**
  Interface card for the IP406 and IP412 providing 2 x T1/PRI (48 lines).
- **IP400 Quad Analog Trunk (Universal) (700359938)**
  Interface card for the IP406 and IP412 providing 4 x Loop start analog trunks. Universal variant supports specifications for North America, Europe and New Zealand.

© 2008 AVAYA All rights reserved.

Country Availability

IP Office is available from distribution partners in the following countries. Please refer to your country price list for the availability of individual items.

Americas	Europe, Middle East and Africa		Asia Pacific
• Argentina • Brazil • Canada • Chile • Colombia • Mexico • Peru • USA	• Austria • Belgium • Bulgaria • Croatia • Cyprus • Czech Republic • Denmark • Estonia • Finland • France • Germany • Greece • Hungary • Iceland • Ireland • Italy	• Latvia • Lithuania • Luxembourg • Malta • Netherlands • Norway • Poland • Portugal • Romania • Russia • Slovenia • Slovakia • South Africa • Spain • Sweden • Switzerland • United Kingdom	• Australia • China • Hong Kong • India • New Zealand • South Korea • Taiwan

© 2008 AVAYA All rights reserved.

Sample Configurations

IP500

Scenario 1:

A US business requiring 190 display phones and 96 digital trunks with 20 analog lines for fallback purposes. This configuration uses an IP500 providing 196 extensions and 96 digital trunks (4 x T1) and two IP400 Analog Trunk 16 modules offering capacity of up to 32 analog trunk lines .

Kit List

- 1 x IP500 control unit.
- 1 x IP500 Feature Key
- 1 x IP Office Standard Edition upgrade to Professional Edition license
- 9 x Region specific power cords.
- 2 x IP500 Digital Station 8 cards
- 2 x IP500 Legacy Card Carriers
- 2 x PRI 48 T1 trunk cards.
- 6 X IP400 Digital Station 30 external expansion modules.
- 2 x IP400 Analog Trunk 16 external expansion modules.
- 190 x Avaya 5410 digital phones.

Scenario 2:

A Business moving to a pure IP Telephony solution with 90 IP hardphones, 90 IP softphones and 60 external trunk lines for its main location and the ability to network with other sites using IP trunking.
This configuration uses an IP500 fitted with a 64-channel Voice Compression Module (VCM). This card allows up to 64 simultaneous calls to external parties (IP extension calling a non-IP telephone or line). For IP to IP calls, VCM resources are only required for initial call set-up. Depending on the typical utilization of external trunks, a lower capacity VCM variant could be employed, as appropriate.

The IP Office softphone is 'Phone Manager Pro PC Softphone' which is an enhanced version of the standard Phone Manager Pro application enabled for each user using two License Keys as listed below.

Kit List

- 1 x IP500 control unit.
- 1 x IP Office Standard Edition upgrade to Professional Edition license
- 1 x IP500 Legacy Card Carrier
- 1 x PRI 60 E1 trunk card.
- 1 x Region specific power cord.
- 1 x IP500 VCM 64 card (4 channels enabled by default).
- 1 x IP500 VCM 60 channel license
- 60 x 5610 IP phones.
- 1 x IP500 Feature Key A-Law
- 1 x IP400 Phone Manager Pro RFA 50.
- 1 x IP400 Phone Manager Pro RFA 10.
- 1 x IP400 Phone Manager PC SoftPhone RFA 50.
- 1 x IP400 Phone Manager PC SoftPhone RFA 10.

© 2008 AVAYA All rights reserved.

IP412

Scenario 1:

A US business requiring 180 display phones and 96 digital trunks with 20 analog lines for fallback purposes. This configuration uses a IP412 providing 180 extensions and 96 digital trunks (4 x T1) and two IP400 Analog Trunk 16 modules offering capacity of up to 32 analog trunk lines . With the addition of a single Dual PRI T1 interface, the system is fitted with an extra trunk card in its spare slot to provide the additional 48 lines.

Kit List

- 1 x IP412 control unit.
- 9 x Region specific power cords.
- 2 x PRI 48 T1 trunk cards.
- 6 X IP400 Digital Station 30 external expansion modules.
- 2 x IP400 Analog Trunk 16 external expansion modules.
- 180 x Avaya 5410 digital phones.

Scenario 2:

A Business moving to a pure IP Telephony solution with 90 IP hardphones, 90 IP softphones and 60 external trunk lines for its main location and the ability to network with other sites using IP trunking.
This configuration uses an IP412 PRI 60 E1 fitted with two 30-channel Voice Compression Modules (VCMs). These two internally fitted cards allow up 60 simultaneous calls to external parties (IP extension calling a non-IP telephone or line). For IP to IP calls, VCM resources are only required for initial call set-up. Depending on the typical utilization of external trunks, a lower capacity VCM variant could be employed, as appropriate.

The IP Office softphone is 'Phone Manager Pro PC Softphone' which is an enhanced version of the standard Phone Manager Pro application enabled for each user using two License Keys as listed below.

Kit List

- 1 x IP412 control unit.
- 1 x PRI 60 E1 trunk card.
- 1 x Region specific power cord.
- 2 x IP400 VCM 30 cards.
- 60 x 5610 IP phones.
- 1 x IP Office Feature Key
- 1 x IP400 Phone Manager Pro RFA 50.
- 1 x IP400 Phone Manager Pro RFA 10.
- 1 x IP400 Phone Manager PC SoftPhone RFA 50.
- 1 x IP400 Phone Manager PC SoftPhone RFA 10.

© 2008 AVAYA All rights reserved.

© 2008 AVAYA All rights reserved.

Chapter 17.
TAPI Functions Supported by IP Office

© 2008 AVAYA All rights reserved.

17. TAPI Functions Supported by IP Office

TAPI 2.1 Functions Supported

TAPI*Link* Lite provides the following functionality for TAPI 2.1:

- lineAddToConference
- lineAnswer
- lineBlindtransfer
- lineCompleteTransfer
- lineConfigDialog
- lineClose
- lineDeallocateCall
- lineDial
- lineDrop
- lineGetAddressCaps
- lineGetAddressID
- lineGetAddressStatus
- lineGetAppPriority
- lineGetCallInfo
- lineGetCallStatus
- lineGetDevCaps
- lineGetID
- lineHold
- lineInitialiseEx
- lineMakeCall
- lineNegotiateTAPIVersion
- lineOpen
- linePark
- lineRedirect
- lineRemoveFromConference
- lineSetAppPriority
- lineSetAppSpecific
- lineSetCallPrivilege
- lineSetStatusMessages
- lineSetupTransfer
- lineShutdown
- lineSwapHold
- lineUnhold
- lineUnpark
- lineSetCallData
- lineDevSpecific
- lineGenerateDigits
- lineGenerateTone
- lineMonitorDigits
- lineMonitorTones

TAPI 3.0 functions supported

The following functions are supported using TAPI 3.0:

- **ITTAPI**
- Initialize
- Shutdown
- EnumerateAddresses
- RegisterCallNotifications
- Put_EventFilter

- **ITAddress**
- get_AddressName
- get_dialableAddress
- get_ServiceProviderName
- CreateCall

- **ITMediaSupport**
- get_MediaTypes

- **ITCallInfo**
- get_Address
- get_CallState
- get_CallInfoString
- SetCallInfoBuffer

- **ITBasicCallControl**
- Connect
- Answer
- Disconnect
- Hold
- SwapHold
- ParkDirect
- Unpark
- BlindTransfer
- Transfer

- **ITCallStateEvent**
- get_Cause
- get_State
- get_Call

- **ITCallNotificationEvent**
- get_Call

- **ITCallInfoChangeEvent**
- get_Call

- **ITCallHubEvent**
- get_Event
- get_Call

Notes:

- TAPI*Link* Lite can be used from C, C++ and Delphi. Visual Basic cannot directly use TAPI 2.1, but does support TAPI 3.0 without any third-party tools.
- TAPI*Link* Lite provides detailed information on telephony events, including the ability to screen-pop based on CLI and/or DDI.

© 2008 AVAYA All rights reserved.

Device-Specific Data

The following table shows the device specific data available via TAPI.

- Phone's extension number
- Forward on busy flag
- Forward on no answer flag
- Forward unconditional flag
- Forward hunt group flag
- Do not disturb flag
- Outgoing call bar flag
- Call waiting on flag
- Voicemail on flag
- Voicemail ring-back flag
- Number of voicemail messages
- Number of unread voicemail messages
- Outside call sequence number
- Inside call sequence number
- Ring back sequence number
- No answer timeout period
- Wrap up time period
- Can intrude flag
- Cannot be intruded upon flag
- X directory flag
- Force login flag
- Login code flag
- System phone flag
- Absent message id
- Absent message set flag
- Voicemail email mode
- User's extension number
- Users Locale
- Forward number
- Follow me number
- Absent text
- Do not disturb exception list
- Forward on busy number
- User's priority
- Number of groups the user is a member of
- Number of groups that the user is a member of that are currently outside their time profile
- Number of groups the user is currently disabled from
- Number of groups that the user is a member of that are currently out of service
- Number of groups that the user is a member of that are currently on night service

© 2008 AVAYA All rights reserved.

DevLink Fields

The following table shows the device specific data available via DevLink.

#	Field Data (S Message)	#	Field Data (S Message)
1	A call id	26	Voicemail disallow
2	B call id	27	Sending complete
3	A state	28	Bc.tc,bc.tm
4	B state	29	Owner hunt group name
5	A connected	30	Original hunt group name
6	A is music	31	Original user name
7	B connected	32	Target hunt group name
8	B is music	33	Target user name
9	A name	34	Target RAS name
10	B name	35	Is internal call
11	B list (possible targets for the call)	36	Time stamp
12	A slot ,channel	37	Connected time
13	B slot , channel	38	Ring time
14	Called party presentation and type	39	Connected duration
15	Called party number	40	Ring duration
16	Calling party presentation and type	41	Locale
17	Calling party number	42	Park slot number
18	Called sub address	43	Call waiting
19	Calling sub address	44	Tag
20	Dialled party type	45	Transferring
21	Dialled party number	46	Sv active
22	Keypad type	47	Sv quota used
23	Keypad number	48	Sv quota time
24	Ring attempt count	49	Account code
25	Cause	50	Unique call identifier
#	**Field Data (D Message)**	**#**	**Field Data (A Message)**
1	A call id	1	A call id
2	B call id	2	B call id
3	Unique call identifier	3	Unique call identifier

© 2008 AVAYA All rights reserved.

© 2008 AVAYA All rights reserved.

Chapter 18.
Technical Specifications

© 2008 AVAYA All rights reserved.

18. Technical Specifications

Dimensions

Unit Dimensions (mm/inches)	Width	Height	Depth
IP406 V2, IP412 and all Expansion Modules	445mm/17.5"	71mm/2.8"	245mm/9.7"
IP Office - Small Office Edition	255mm/10.0"	76mm/3.0"	241mm/9.5"
IP500	445mmm/17.5"	73mm/2.9"	365mm/14.4"

- The recommended minimum clearance, front and rear, for the connection of cables and other devices is 75mm/3".

Weight

Unit	Weight
IP500 System Unit	3.2Kg/7.0lbs
IP406 V2 Control Unit	3.0Kg/6.7lbs
IP412 Control Unit	3.0Kg/6.7lbs
Small Office Edition	1.2Kg/2.6lbs
Analog 16 Module	2.9Kg/6.5lbs
DS16 Module	3.0Kg/6.7lbs
DS30 Module	3.5Kg/7.8lbs
WAN3 Module	2.8Kg/6.3lbs
So8 Module	2.8Kg/6.3lbs
Phone 8 Module	2.8Kg/6.3lbs
Phone 16 Module	2.9Kg/6.5lbs
Phone 30 Module	3.1Kg/6.94lbs

Environmental

- 0°C to +40°C (32°F to 104°F). 95% relative humidity, non-condensing.

© 2008 AVAYA All rights reserved.

Telephone Extension Cable Lengths

The following table details the maximum cable lengths supported for the telephone ranges. These figures assume that standard twisted-pair telephone cable or CAT5 network cable is used.

Telephone	Unshielded Twisted-Pair (UTP) - 50nf/Km			CW1308
	AWG22 (0.65mm)	AWG24 (0.5mm)	AWG26 (0.4mm)	
2400/5400 Series	1200m/3937'.	1000m/3280'.	670m/2200'.	400m/1310'.
4406D Phone	1000m/3280'.	1000m/3280'.	400m/1310'.	400m/1310'.
4412D Phone	1000m/3280'.	700m/2295'.	400m/1310'.	400m/1310'.
4424D	500m/1640'.	500m/1640'.	400m/1310'.	400m/1310'.
6400 Series	1000m/3280'.	1000m/3280'.	400m/1310'.	400m/1310'.
T3 Series (Upn)	1000m/3280'.	1000m/3280'.	400m/1310'.	–
Analog Phones	1000m/3280'.	1000m/ 3280'.	400m/1640'.	800m/2620'.

Heat Dissipation

Note that the above numbers are for reference only. For practical purposes, for example the calculation of heat dissipation, it is recommended to base environmental requirements (for example air cooling or UPS ratings) on the maximum input rating of the power supplies of the planned IP Office configuration, as follows.

In order to calculate the maximum, that is worst case, amount of heat that can be generated by an IP Office system, it is assumed that all input power is converted to heat; whether from the PSU itself, the system unit, expansion module and/or cabling.

Heat dissipation is normally measured in British Thermal Units (BTU's). A heat value expressed in Watts can be converted to BTU/hr by multiplying by 3.41297. As indicated above, you should use the maximum power input of 115 VA of each power supply to calculate this most accurately.

Using the conversion factor:

- Heat Dissipation = 115 x 3.41297 = 392.5 BTU/hour.

The metric equivalent to BTU is a Joule where 1 BTU = 1,055 Joules.

This calculates the BTU value per power supply. The maximum BTU per system is therefore calculated, based on total number of power supplies installed in the system. For example, for a IP412, this would be 1 for the base unit and up to 12 for the expansion modules.

- IP412 Maximum Heat Dissipation = 13 x 392.5 = 5,103 BTU/hr.

Remember to budget for the power requirements of any additional devices that are to be co-located with the IP Office such as server PC's (voicemail, etc).

Power Supply

- **Input**
  - **Small Office Edition:** 2.5mm DC inlet socket. 24Vdc power input. Rating 24V DC, 1.8A maximum.
  - **IP406 V2, IP412 and expansion modules:** 2.5mm DC inlet socket. 24Vdc power input. Rating 24V DC, 2A maximum.
  - **IP Office 500 System Unit:** IEC AC inlet socket. 100-240V AC, 50/60Hz, 81-115VA, 2.5A maximum.

- **Power Supply Units:** All CE/UL/Dentori Safety Approved.
  - **Standard 40W Power Supply Unit** (All control and expansion units unless otherwise indicated) Supplied with the control or expansion unit. 40W PSU with integral lead to the unit. Connection to switched mains supply requires separately supplied country specific IEC 60320 C7 power cord (2-wire figure 8 connector).
    - **Input:** 100-240V AC, 50/60Hz, 81-115VA, 2A maximum.
    - **Output:** 24Vdc, 1.875A, output power 45W maximum.

© 2008 AVAYA All rights reserved.

- **Small Office 45W Power Supply Unit**
  Supplied with the unit. 45W PSU with integral lead to control unit. Connection to switched mains supply requires separately supplied country specific IEC 60320 C13 power cord (3-wire earthed cold kettle lead).
  - **Input:** 100-240V AC, 50/60Hz, 81-115VA, 1.5A maximum.
  - **Output:** 24V DC, 1.875A, output power 45W maximum.
- **IP406 V2 60W Power Supply Unit**
  Supplied with the control or expansion unit. 60W PSU with integral lead to the unit. Connection to switched mains supply requires separately supplied country specific IEC 60320 C13 power cord (3-wire earthed cold kettle lead).
  - **Input:** 100-240V AC, 50/60Hz, 81-115VA, 2.5A maximum.
  - **Output:** 24V DC, 1.5A, output power 60W maximum.
- **IP Office 500 80W internal Power Supply**
  Integral to the System Unit. Connection to switched mains supply requires separately supplied country specific IEC 60320 C13 power cord (3-wire earthed cold kettle lead).
  - **Input:** 100-240V AC, 50/60Hz, 81-115VA, 2.5A maximum.

© 2008 AVAYA All rights reserved.

Interfaces

Interface	Information
DTE Port	• 25 way D-Type female connector, V.24/V.28. • 9 way D-type on IP412, IP406 V2, IP500 and IP Office - Small Office Edition.
ISDN Ports	**EU Interfaces:** • BRI: RJ45 sockets. ETSI T-Bus Interface to CTR3 for Pan European Connection. • PRI E1: RJ45 socket. ETSI T-Bus Interface to CTR4 for Pan European Connection. • PRI T1/J1: RJ45 socket: FCC Part 68/JATE connection. **USA Interfaces:** • PRI T1 Service: Ground Start (GS) – Default, E&M, 56k data for 5ESS, 56/64/64 restricted for 4ESS. • PRI ISDN Switch support: 4ESS, 5ESS, DMS-100, DMS-250 (includes conformance to ANSI T1.607 and Bellcore Special Report SR4287, 1992). • PRI ISDN Services: AT&T Megacom 800, AT&T WATS (4ESS), AT&T SDS Accunet 56kB/s and 64kB/s (4ESS), AT&T Multiquest (4ESS).
Analog Trunk Ports	• RJ45 sockets: Loop start/Ground start (regional dependant)
Power Fail Ports	• RJ45 sockets:
ISDN Data Rates	• BRI: B-channel 64kbps or 56kbps, D-channel 16kbps. • PRI: B-channel 64kbps or 56kbps, D-channel 64kbps.
Analog Phone Ports	• RJ45 sockets: • CLI Schemes: DTMFA, DTMFC, DTMFD, FSK and UK20. • REN: 2. (External Bell via POT port: REN = 1) • Off Hook Current: 25mA. • Ring Voltage: 40V (nominal) RMS.
LAN	• RJ45 sockets. Auto-negotiating 10/100 BaseT Ethernet (10/100Mbps).
WAN	• Small Office Edition: RJ45 Ethernet socket. • IP406 V2 and IP412 (optional on Small Office Edition): 37 way D-Type female sockets. X.21 interface to 2048k bps, V.35 interface to 2048Kbps and V.24 Interface to 19.2Kbps.
Audio	• 3.5mm Stereo Jack socket. Input impedance - 10k /channel. • Maximum AC signal – 200mV rms.
External Output Port	• 3.5mm Stereo Jack socket. Switching Capacity - 0.7A. • Maximum Voltage - 55V DC. On state resistance - 0.7. • Short circuit current - 1A. Reverse circuit current capacity - 1.4A.
Wireless Module	• Small Office Edition only. • 16bit Type II PCMCIA format PC card. • IEEE 802.11b WiFi.
Embedded Voice Memory	• Small Office Edition: 64MB Flash memory, 16bit Type II PCMCIA card. • IP406 V2 and IP500: 512MB Compact Flash memory card.

© 2008 AVAYA All rights reserved.

Protocols

Protocol	RFC	Information
V120	-	A standard Rate Adaptation mechanism.
V110	-	A standard Rate Adaptation mechanism.
PPP	RFC1661	Point to Point Protocol.
LCP	RFC1570	Link Control Protocol.
MP	RFC1990	Multi-Link (Point to Point) Protocol.
IPCP	RFC1332	Internet Protocol Control Protocol.
PAP	RFC1334	Password Authentication Protocol.
RTP/RTCP	RFC1889	Real Time and Real Time Control Protocol.
CHAP	RFC1994	Challenge Handshake Authentication Protocol.
CCP	RFC1962	Compression Control Protocol.
STAC	RFC1974	STAC LZS Compression Protocol.
MPPC	RFC2118	Microsoft Point to Point Compression (Protocol).
BACP	RFC2125	Bandwidth Allocation Control Protocol.
UDP	RFC768	User Datagram Protocol.
IP	RFC791	Internet Protocol.
TCP	RFC793	Transmission Control Protocol.
DHCP	RFC1533	Dynamic Host Control Protocol.
NAT	RFC1631	Network Address Translation.
BOOTP	RFC951	Bootstrap Protocol.
TFTP	RFC1350	Trivial File Transfer Protocol.
NTP	RFC868	Network Time Protocol.
SNMPv1	RFC1157	Simple Network Management Protocol. (STD15)
	RFC1155	Structure and identification of management information for TCP/IP based internets. (STD16)
	RFC1212	Concise MIB Definitions. (STD16)
	RFC1215	A convention for defining traps for use with SNMP.
MIB-II	RFC1213	Managment Information base for network management of TCP/IP based internets: MIB-II. (STD17)
ENTITY MIB	RFC2737	Entity MIB (Version 2).
RIP	RFC1058	Routing Information Protocol.
	RFC2453	RIP Version 2. (STD56)
	RFC1722	RIP Version 2 Protocol Applicability Statement. (STD57)
IPSec	RFC2401	Security Architecture for the Internet Protocol.
	RFC2402	IP Authentication Header.
	RFC2403	The Use of HMAC-MD5-96 within ESP and AH.
	RFC2404	The Use of HMAC-SHA-1-96 within ESP and AH.
	RFC2405	The ESP DES-CBC Cipher Algorithm with Explicit IV.
	RFC2406	IP Encapsulation Security Payload. (ESP)
	RFC2407	The Internet IP Security Domain of Interpolation for ISAKMP.
	RFC2408	Internet Security Association and Key Management Protocol.
	RFC2409	The Internet Key Exchange.
	RFC2410	The NULL Encryption Algorithm and its Use with IPSec.
	RFC2411	IP Security Document Roadmap.
L2TP	RFC2661	Layer Two Tunneling Protocol "L2TP".
	RFC3193	Securing L2TP using IPSec.
Header Compression	RFC2507	IP Header Compression (IPHC).
	RFC2508	Compressing IP/UDP/RTP Headers for Low-Speed Serial Links.
	RFC2509	IP Header Compression over PPP.
DiffServ	RFC2474	Definition of the Differentiated Services Field (DS Field) in the IPv4 and IPv6 Headers.
PPP MP	RFC1990	The PPP Multilink Protocol (MP).
Frame Relay Encapsulation	RFC1490	Multi protocol Interconnect over Frame Relay.
ML-PPP	RFC2686	The Multi-Class Extension to Multi-Link PPP.

© 2008 AVAYA All rights reserved.

Session Initiation Protocol

- Rec. E.164 [2] - ITU-T Recommendation E.164: The international public telecommunication numbering plan
- RFC 2833 [7] - RTP Payload for DTMF Digits, Telephony Tones and Telephony Signals
- RFC 3261 [8] - SIP: Session Initiation Protocol
- RFC 3263 [10] - Session Initiation Protocol (SIP): Locating SIP Servers
- RFC 3264 [11] - An Offer/Answer Model with Session Description Protocol (SDP)
- RFC 3323 [14] - A Privacy Mechanism for the Session Initiation Protocol (SIP)
- RFC 3489 [18] - STUN - Simple Traversal of User Datagram Protocol (UDP) Through Network Address Translators (NATs)
- RFC 3824 [24] - Using E.164 numbers with the Session Initiation Protocol (SIP)

© 2008 AVAYA All rights reserved.

© 2008 AVAYA All rights reserved.

Chapter 19. Appendix

© 2008 AVAYA All rights reserved.

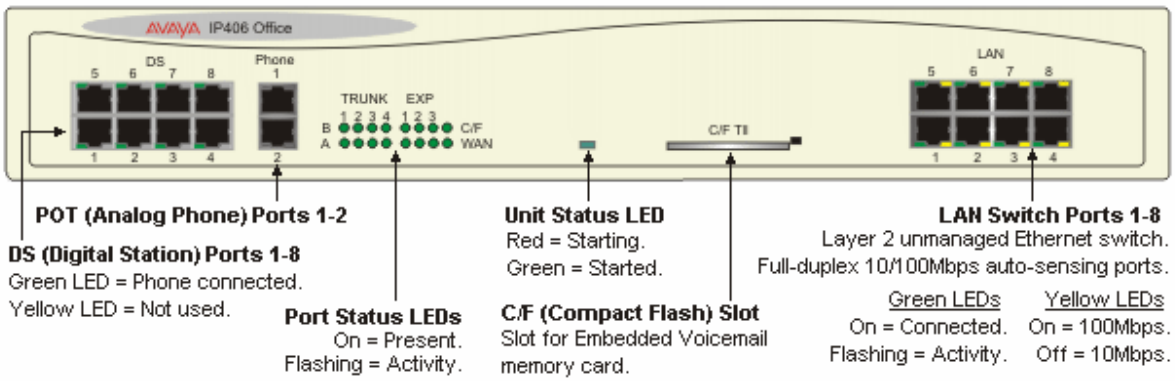
19. Appendix

The following equipment is no longer sold by Avaya but is still supported by IP Office 4.2.

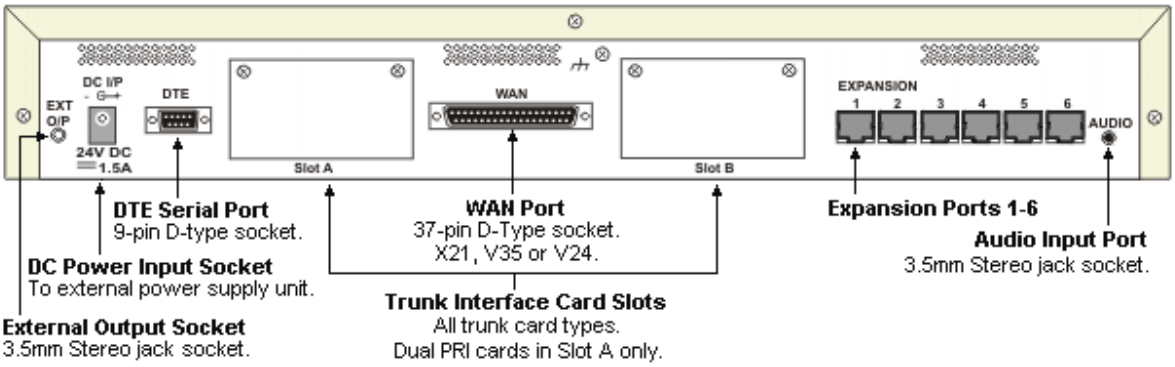
IP406 V2 Control Unit

The IP406 V2 control unit is a stackable unit with an optional 19" rack mounting kit. The IP406 V2 includes:

- Eight Digital Station (DS) ports for supported 2400, 4400, 5400, 6400 and T3 Series phones plus 3810 wireless (US) phones.
- Two Analog telephone ports.
  - Two Wire
  - DTMF signaling (No rotary or Loop Disconnect)
  - Timed Break Recall (No Earth Loop Recall)
  - Caller ID capable – a variety of standards, see later
  - MWI capable – 82.5V and Line Reversal
- Eight 10/100 Mbps LAN Switched ports (Layer-2, unmanaged).
- Support for optional embedded voicemail/auto-attendant (Compact Flash card)
- 9-pin DTE Port (for maintenance or Feature Key connection for application licensing).
- X.21/V35 WAN interface.
- Support for up to 6 IP Office Expansion Modules:
  - Phone modules (8, 16, 30)
  - Digital Station modules (16, 30)
  - Analog Trunk Module 16
  - So8 module
- External O/P socket supporting two relay on/off switch ports, e.g. for door entry systems.
- Audio input port for external music on hold source.
- Two trunk interface card slots for analog, BRI, PRI (T1, E1) or CAS (E1R2)
- Internal socket for IP Telephony expansion – voice compression modules (from 4 to 30 channels)
- Internal socket for internal modem (2 or 12) for Remote Access Services
- 50 Data channels
- Up to 20 VoiceMail Pro ports



© 2008 AVAYA All rights reserved.



DTE Serial Port
9-pin D-type socket.

DC Power Input Socket
To external power supply unit.

External Output Socket
3.5mm Stereo jack socket.

WAN Port
37-pin D-Type socket.
X21, V35 or V24.

Trunk Interface Card Slots
All trunk card types.
Dual PRI cards in Slot A only.

Expansion Ports 1-6

Audio Input Port
3.5mm Stereo jack socket.

© 2008 AVAYA All rights reserved.

Expansion Modules

Through support of up to six external Expansion Modules, IP406 can be enhanced to support a mixture of analog, digital or IP phones, to maximum of 190 phones in any combination.

If additional analog trunks are required, these can be aggregated in groups of 16 on each analog expansion module

Data Channels

A Data Channel is used for Remote Access (RAS), Internet Access, and Voicemail sessions. A data channel is an internal signaling resource used whenever a call is made from the IP network to an exchange line (Central Office). For example, four people surfing the Internet will use a single data channel since they all share the same line to the ISP. Two people remotely accessing the Office LAN from home will use two data channels since they have dialed in on separate lines. IP extensions do not use data channels. Data channels are used for voicemail connections with a maximum of 20 available for VoiceMail Pro on an IP406 V2.

Modems and Voice Compression modules

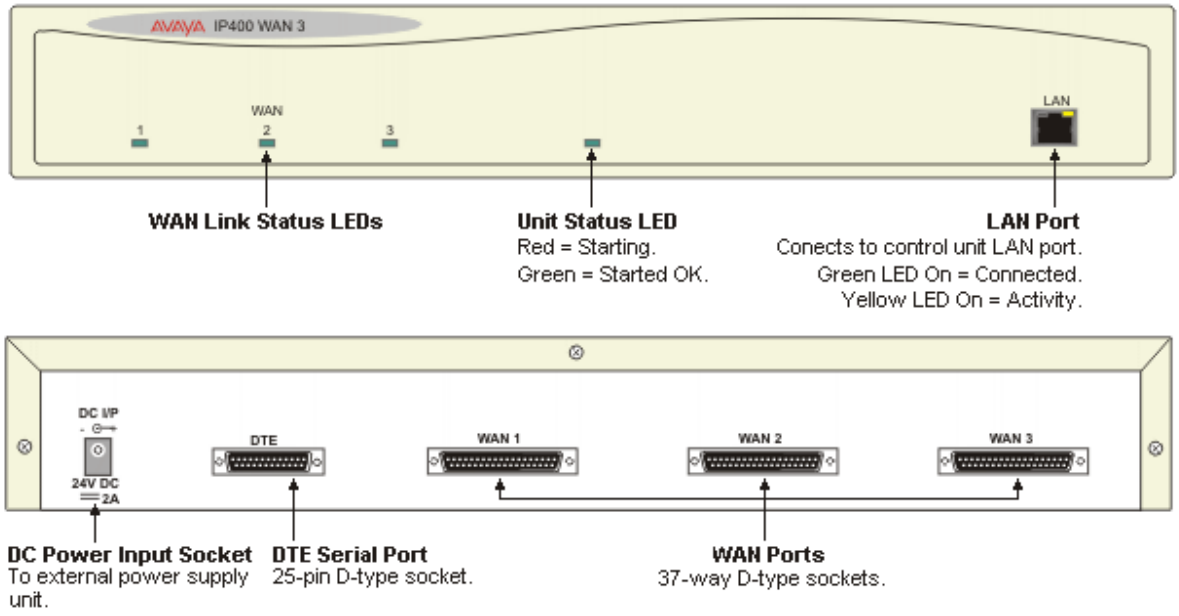
You can add additional hardware to the IP406 system to add one modem card (2 or 12 V.90 modems) and 1 Voice Compression Module (VCM). The VCM supports from 4 to 30 simultaneous Voice over IP sessions and is used for either providing networking between sites over a Wide Area Network or supporting IP Telephones and Soft phones.

© 2008 AVAYA All rights reserved.

IP400 WAN3 10/100

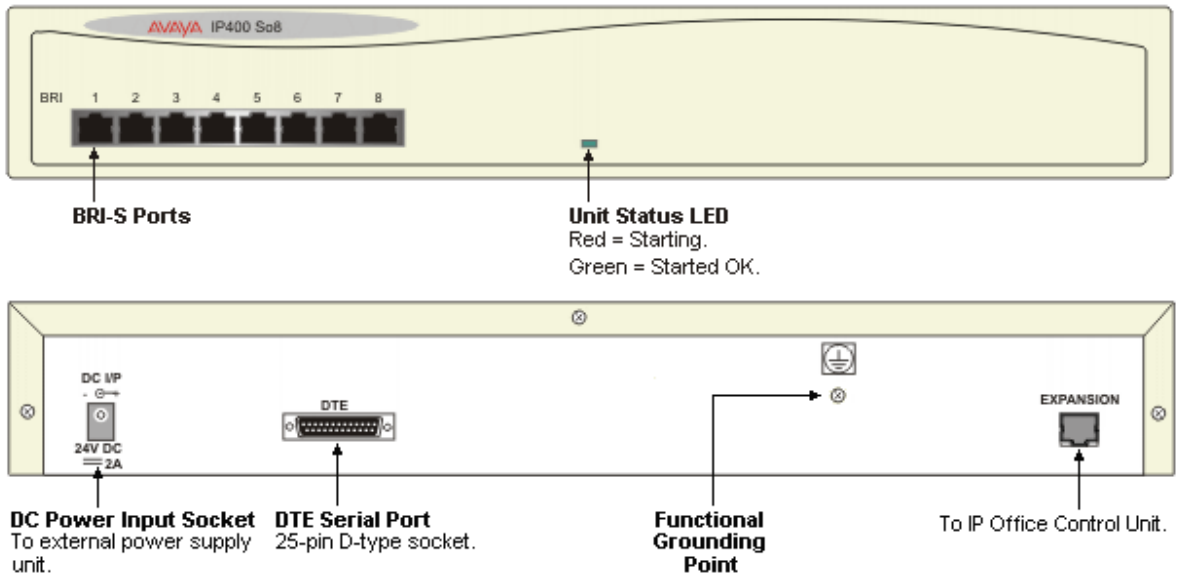
The IP400 WAN3 10/100 module provides three WAN connections (X21, V35 or V24 via a 37way D Type socket and using an appropriate connector cable). Data rates of up to 2 Mbps are supported on each interface, the carrier providing the line dictates the actual operating speed i.e. in some territories the maximum speed may be limited to 1.544 Mbps. These WAN interfaces are identical to the single WAN connection provided as standard on the IP406 and IP412 platforms.

The IP400 WAN3 10/100 connects to the control unit through the Local Area Network via a 10/100Mbps connection and does not use an expansion port on the control unit. Small Office Edition and IP500 do not support the WAN3 10/100. All other platforms support up to two WAN3 10/100 modules.



IP400 So8 Expansion Module

The IP400 So8 module provides 8 S-Bus interfaces for Basic Rate ISDN devices, such as video conferencing, fax servers or ISDN telephones.



The IP Office So8 expansion module supports both point-to-point and point-to-multipoint connections. A maximum of 10 terminal endpoints identifiers (TEIs) are supported on each bus.

© 2008 AVAYA All rights reserved.

Small Office Edition

The IP Office - Small Office Edition is the entry level control unit of the IP Office solution and is delivered in a compact configuration that provides a mix of Analog trunks, Analog and Digital extensions and Voice over IP (VoIP) capacity. Dependant on the model chosen, up to a maximum of 28 extensions can be supported (4 Analog, 8 Digital and 16 IP).

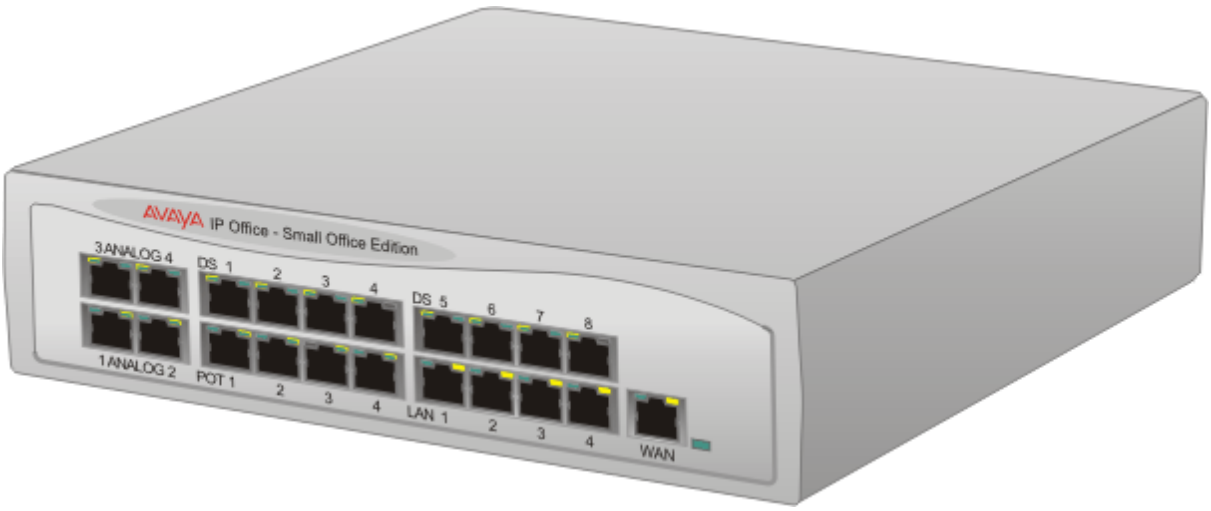
All IP Office - Small Office Edition's variants have a four-port Layer 2 Ethernet Switch and a dedicated switched Ethernet WAN port (Layer 3), making the system ideal for connection to local area networks and broadband wide area network services such as ADSL and Cable. With Voice over IP as standard and optional IPSec security, the system can be quickly configured to provide secure voice and data networking from remote offices or branch locations back to a head office over a broadband connection.

The IP Office - Small Office Edition includes a WAN option slot on the rear of the unit which can be used to support other network connection types such as V35, V24, X21 and T1 leased lines.

The back of the unit also features a twin PCMCIA socket that can support a plug-in voice memory card for use with the embedded voicemail function, and a Wireless LAN card when using the system as an Access Point.

To enable licensed IP Office applications, a serial Feature Key can be attached directly to the IP Office - Small Office Edition removing the need for an external PC for license verification.

For resilience, under power fail conditions Analog trunk port 2 is connected to POT extension port 1.



The pre-defined configurations supported in IP Office 3.1 are detailed in the following table.

IP Office - Small Office Edition	Analog Trunks	Analog Extensions	Digital Stations	IP Extensions	VoIP Channels
4T+4A+8DS (3 VoIP)	4	4	8	16	3
4T+4A+8DS (16 VoIP)	4	4	8	16	16

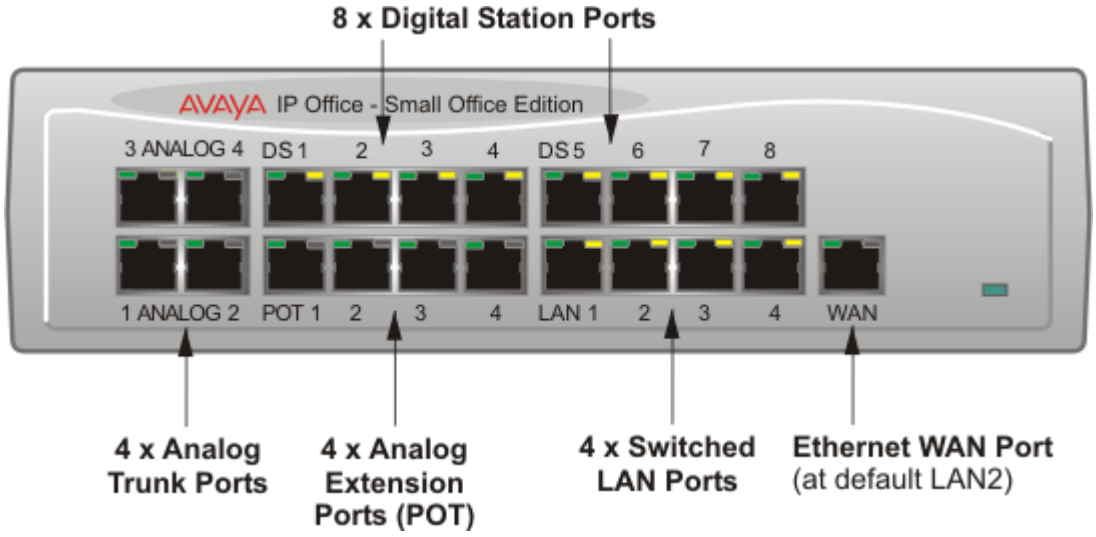
- During power fail, Analog port 2 is connected to POT port 1.

© 2008 AVAYA All rights reserved.

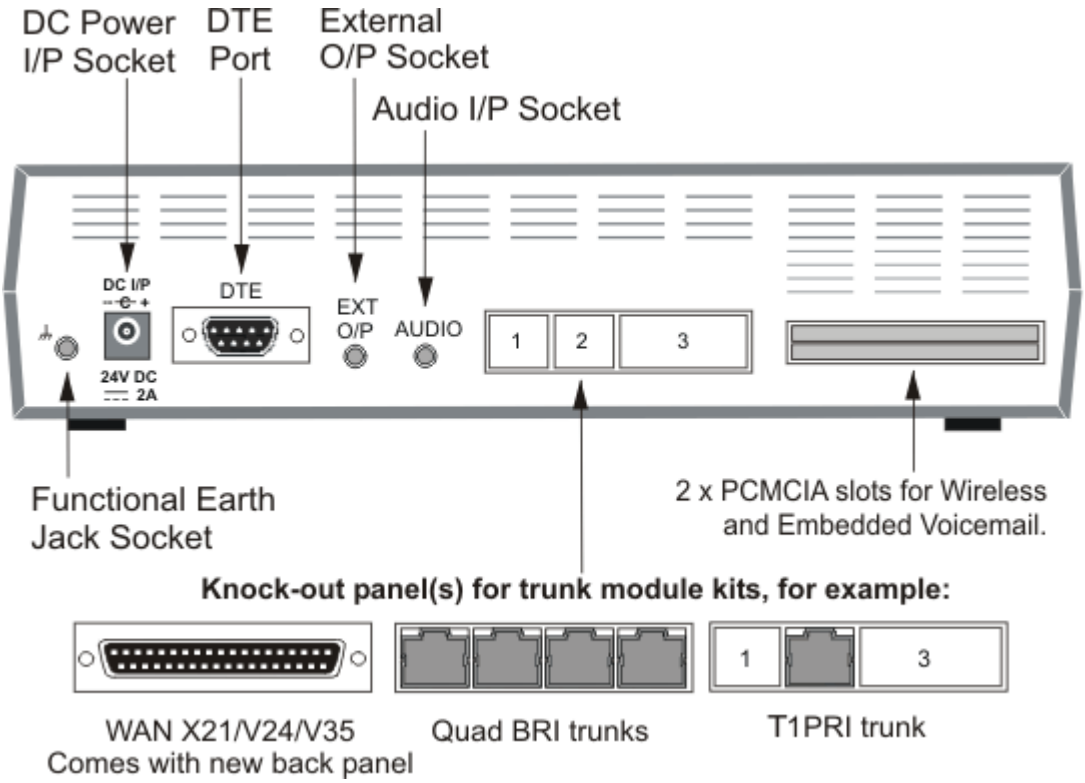
IP Office - Small Office Edition 4T+4A+8DS (3 or 16 VoIP)

The IP Office - Small Office Edition 4T+4A+8DS provides:

- Four Analog Loop Start Trunks (Caller ID enabled).
- Four Analog extension (POT) ports with power fail switchover such that analog trunk port 2 is connected to analog extension port 1.
- Eight Digital Station (DS) ports for selected 2400, 5400 and 6400 phones plus 3810 wireless (US) phones. T3 Series phones are not supported from Release 4.0 onwards on Small Office Edition.
- 3 or 16 VoIP Codecs (G.723.1, G.711 and G.729a) and 48ms echo cancellation.
- 4 Switched Ethernet ports (Layer 2).
- Dedicated Switched Ethernet WAN port (Layer 3).
- 2 x PCMCIA Slots for optional Wireless and Embedded Voicemail card support.
- Expansion Slot for optional WAN card (V35/V24/X.21), BRI or T1 PRI.
- Serial DTE port.
- Audio input port for external music on hold source.
- External O/P socket supporting two relay on/off switch ports, e.g. for door entry systems.



© 2008 AVAYA All rights reserved.



© 2008 AVAYA All rights reserved.

IP Office - Small Office Edition WAN Expansion Interfaces

Both IP Office - Small Office Edition variants provide an expansion slot for an optional WAN interface of the following types (check locally for availability). Each of these interface cards are now described in more detail.

IP400 WAN Expansion

The IP400 WAN Expansion card provides a single WAN connection (X21, V24 or V35 via a 37-way D Type socket). Line speeds up to and including 2Mbps are supported on the interface. The carrier providing the line dictates the actual operating speed, i.e. in some territories the maximum speed may be 1.544M.

IP400 BRI Card

The BRI trunk card provides 4 European Basic Rate ISDN T interfaces (8 trunks).

Details of the supported supplementary services on BRI interfaces are given in the 'Public and Private Voice Networks' section.

- Not available in all territories, check for availability.

IP400 T1 PRI Card

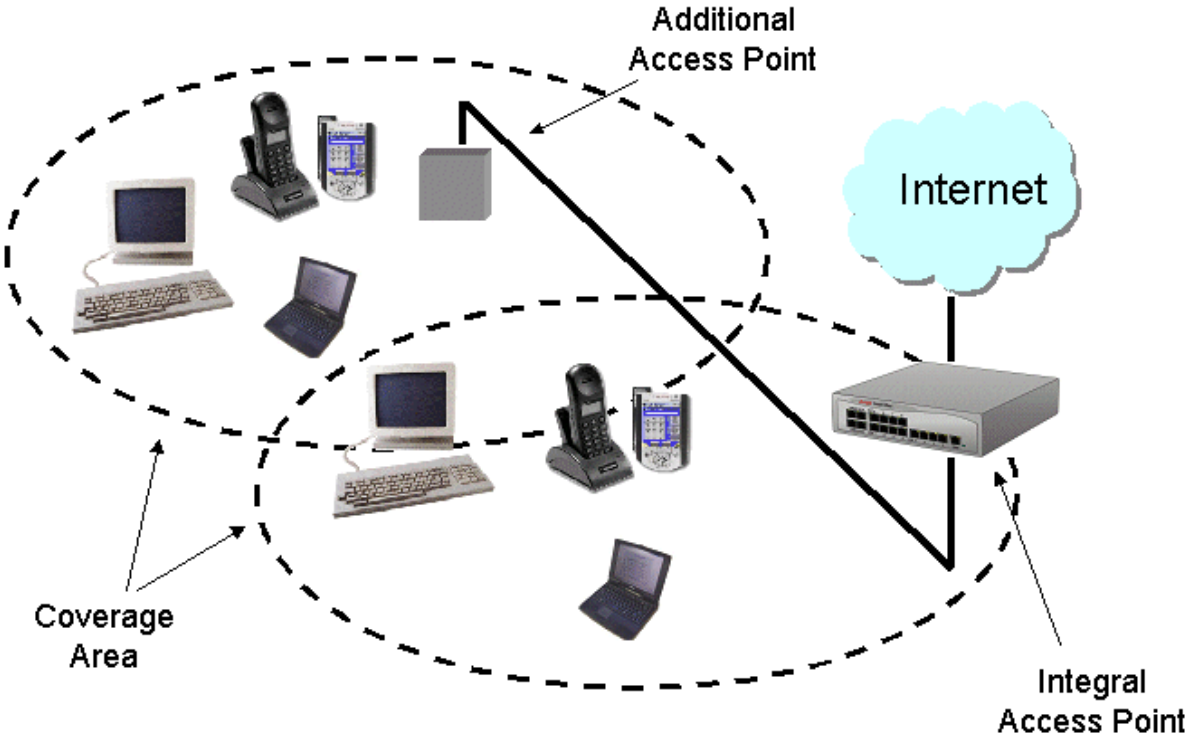
The IP400 T1 PRI card provides a single primary rate trunk interface for supporting voice services and fractional leased lines, providing up to 256K bandwidth on IP and Frame Relay services.

- Not available in all territories, check for availability.

© 2008 AVAYA All rights reserved.

Optional Wireless Access Point

All IP Office - Small Office Edition platforms can be configured to become Wireless LAN access points. An Access Point acts as a Hub in a wireless network providing connectivity between devices in the vicinity. In ideal conditions a range of up to 550M (1750 ft) is achievable although this range will be decreased if walls and other obstacles are present. This is used where local conditions impair coverage and additional Access Points are needed to cover the black spots.



The IP Office - Small Office Edition wireless network can be secured against intruders using either the Wired Equivalent Privacy (WEP) or RC4. WEP uses 64 bit encryption key and RC4 uses a 128 bit encryption key. Only devices with a matching security key can participate in the network.

IP Office - Small Office Edition complies to the IEEE 802.11 and IEEE 802.11b standards meeting the Wireless Ethernet Compatibility Alliance (WECA) Wireless Fidelity Wi-Fi requirements for interoperability.

Summary

- 2.4 GHz to 2.5 GHz band (Scientific Medical and Industrial (SMI) band).
- Automatic fallback 11Mbps, 5.5Mbps, 2Mbps or 1Mbps.
- IEEE 802.11 and IEEE 802.11b Compliance.
- Wireless Fidelity Wi-Fi Compliance.
- Interoperable with other 802.11b compliant devices.
- WEP or RC4 security.
- Range up to 550M (1750ft).

Range (meters/ft)	11Mbps	5.5Mbps	2Mbps	1Mbps
Open	160m/252ft	270m/885ft	400m/1300ft	550m/1750ft
Semi-Open	50m/165ft	70m/230ft	90m/300ft	115m/375ft
Closed	25m/80ft	35m/115ft	40m/130ft	50m/165ft
Receiver Sensitivity dBm	-82	-87	-91	-94
Delay Spread (at FER of <1%)	65ns	225ns	400ns	500ns

For wireless operation, IP Office - Small Office Edition must be fitted with a Wireless LAN card and the Wireless LAN Access Point license key. Alternatively, a 3rd party wireless access point can be connected directly to one of the LAN ports.

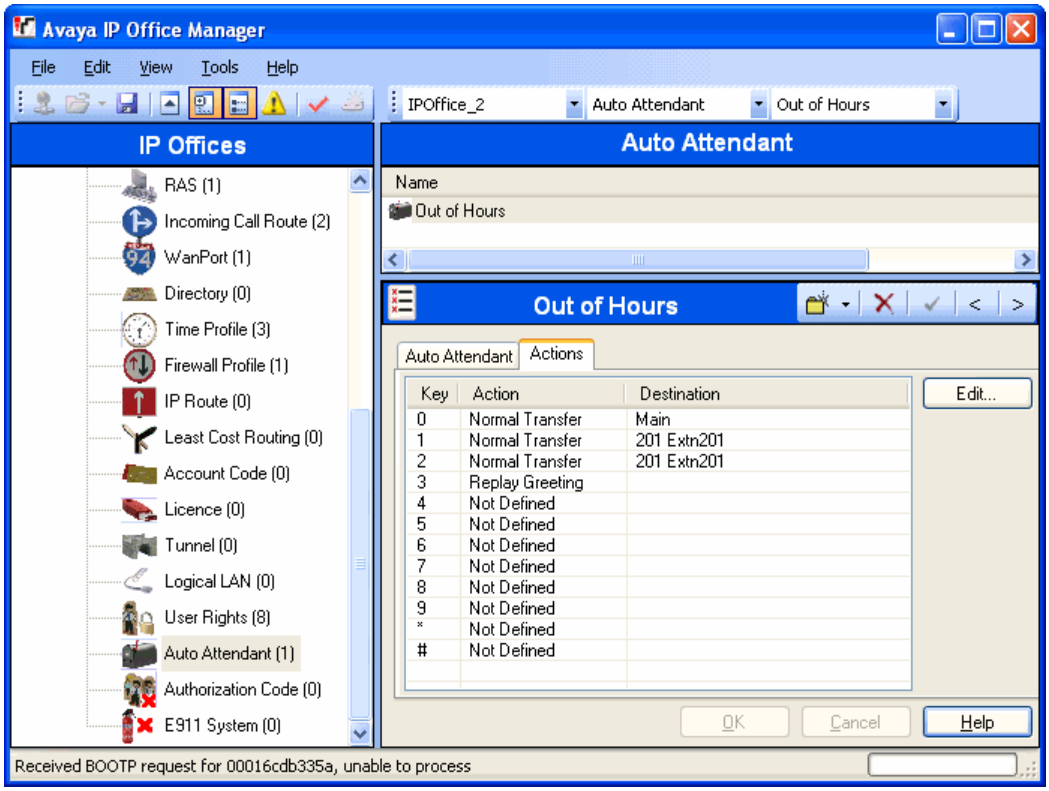
© 2008 AVAYA All rights reserved.

Optional Embedded Voicemail with Auto-Attendant

Entry-level voicemail and auto attendant applications are available using the Avaya memory expansion kit in one of the PCMCIA slots on the rear of the Small Office Edition. This provides small locations with an effective embedded messaging solution with auto-attendant without the additional costs of an external PC. The embedded voicemail supports up to 10 hours of message storage. The number of available voicemail ports (to support simultaneous calls to voicemail) is 3 ports on the 3 VoIP model or 10 ports on the 16 VoIP model

Personalized greetings and PIN-code access can be enabled for each mailbox by the mailbox users. Inactivity timeout and return to operator options ensure efficient message handling. Mailbox users can also access their mailboxes when out of office using a simple remote login sequence.

Up to 40 independent auto-attendants can be configured on the platform. These may be linked together to form multiple tiers of attendants. The choice of which auto-attendant is to answer a call can be made on any of the criteria on the Incoming Call Routing form such as called number, calling number and time of day.



Each auto-attendant has a single menu of 12 items (0…9, *, #) that a caller can select from to either be transferred to a predefined number, to another auto attendant or to replay the greeting. The greeting for the menu is controlled by time profiles to allow three alternative messages to be played i.e. Morning, Afternoon and Evening. These messages can be labeled and then re-used in as many auto-attendants as required.

Please note that the IP406 V2 and IP500 embedded voicemail memory cards are identical but these are not interchangeable with the Small Office Edition card. Only Avaya supplied memory cards with the voicemail and auto attendant applications pre-installed can be used.

© 2008 AVAYA All rights reserved.

© 2008 AVAYA All rights reserved.

Index

- A -

Absence Text 104
  setting 104
  strings 104
Absent Text Message 152
Acceptable Delay 135
Access Point 72, 181
accessing
  Database Information within Call Flows 209
  Email 206
  Office LAN 32
Account 10, 20, 113, 135, 142, 168, 173, 176, 194, 197, 205, 216, 223, 253, 261, 271, 272, 288
Account Activity 253
Account Code Costing Log
  Outgoing 253
Account Code Log
  Outgoing 253
Account Code Recording 216
Account Codes 10, 113, 216, 253, 271, 288
  view 176
Account Codes tab 176
Account Service Report 253
ACD 122
ACM 22, 154, 218
ACM RFA 218
Acquire Call 105
  executing 122
  waiting 122
ACSII-CSV 266
Active Directory 205
ActiveX Data Objects 209
Add/Update Conference Participants 232
Additional ISDN 142
Additionally Music On Hold 232
addressing
  Domain Name Service 164
  voicemails 219
admin 208, 272
Admin portion 272
Administration 6, 46, 117, 140, 173, 190, 205
Administrator 10, 20, 105, 118, 176, 208, 209, 210, 214, 217, 232, 266, 273
  Change 190
ADMM 69
ADO 209
Advanced Small Community 120, 152
Advanced Small Community Networking 120, 152
Advice Of Charge 142, 173, 180
Alarm Calls 219
Alarm Handling 252
Alarm Reporter 251
Alert 56, 58, 60, 69, 82, 106, 111, 112, 119, 130, 173, 176, 184, 202, 265
  User 10, 123
ALG 146
Algorithmic Delay 135
All Calls 10, 112, 176, 251
All-in-one 6, 21
allocated 164, 167, 197, 231, 232, 272
  User Rights 114
Alpha 65
Alphabetic Keystrokes 189
Alphanumeric 64, 75, 91, 176, 219
Alphanumeric Data Collection 219
Alternate Call Routing 18
Alternate Route Selection 106, 114, 157
Analog 16 21, 291
Analog 16 Module 291
Analog Extensions 10, 27, 41, 94, 106, 123
analog lines 230, 282, 283
Analog Phone 21, 22, 30, 113, 124, 128, 291, 294
Analog Phone Ports 294
Analog Telephone Features 90
Analog Telephones 41, 90, 97, 124, 130
Analog Telephones/POTS 89
Analog Trunk 16 277
Analog Trunk 16 EU 277
Analog Trunk 16 NZ 277
Analog Trunk 16-port 142
Analog Trunk Module 25, 32
Analog Trunk Module 16 25, 32
Analog Trunk Ports 294
Analog Trunk Restriction 230
Analog Trunks 25, 30, 32, 34, 114, 142, 157, 163, 230, 277, 282, 283, 294
Analog/digital 134
ANI 142, 180, 217
ANLG 27
Announcements 6, 10, 120, 121, 197, 199, 200, 219, 232, 251
  Queuing 258
ANSI T1.401
  conform 142
ANSI T1.607 294
Answer Bar 261
Answer Interval 115
Answered Calls 184, 257
Anti-Tromboning 152
AOC 125, 142
APAC 68, 69, 91, 97
Appearance 10, 94, 109, 111, 112, 189
Appearance Buttons 111, 112
Appearance Keys
  Twinning 10
appearance/feature 46, 62, 85, 86, 87
Applicable 30, 109, 118, 212, 216

© 2008 AVAYA All rights reserved.

Application Level Gateway 146
Applications Platform 20, 22
Applications Platform Features 20
ARP
receiving 167
ARS 106, 114, 157
Assisted Transfer 200, 219
Audio 10, 25, 32, 64, 65, 66, 72, 106, 130, 131, 135, 137, 179, 200, 217, 228, 229, 232, 237, 277, 294
Audio Codec 135
Audio CODECs 137
Audio Conferencing 228
Audio Volume 64, 65, 66
Audio waveform 217
audio-conference 237
Audiotex 200
Audit 136, 266
AUDIX RFA 154
August 34, 154
August 2003 154
Australia 281
Austria 281
Authorization Codes 113
Auto Attendant Properties
Transfer 10
Auto Attendants on Embedded Voicemail 10
Auto Connect 167
Auto Connect Time Profile 167
auto-attendant offers 208
Auto-Attendant/Audiotex 219
Auto-Create Extensions 131
Automatic Call Distribution 122
Automatic Callback 101
Automatic Intercom 109
Automatic IP 19
Automatic Number Identification 142
automatic/manual
allow 216
automatic/manual recording
calls 216

- B -

Back When Free 125
Backlight 60, 62, 78
backlit 72, 94
BACP 166, 295
Band DTMF 131, 137
Bandwidth Allocation Control Protocol 166, 295
Bandwidth Required For Each 135
Bandwidth Required For Each Voice Call 135
bar 114, 173, 287
queue panel displays 184
Base 6, 10, 22, 25, 27, 30, 46, 69, 71, 82, 94, 164, 277, 291, 295
Base Unit 82, 94, 164, 277, 291
Base Unit Power Supply Adapter 82
BaseT Ethernet 56, 58, 60, 64, 65, 66
Basic Commands 219
Basic Rate 34, 39, 142, 164
Basic Rate ISDN 34, 39
BCC 271
B-channel 64kbps 294
Bearer Capability Class 271
Bellcore Special Report SR4287 294
Belt Clip 72, 80, 81, 82
bi-directional 128
BLF 152, 173, 181, 189, 190, 215, 252
form 117
groups 184
BLF Details 252
BLF Groups 189
BLF Panel 184, 190
Blind Transfer 102
BlindTransfer 286
Book Shop 210
BOOTP 295
Bootstrap Protocol 295
Bothway 167
bps 135
branch maximizes 140
branches 6, 128, 140
DEFINITY/ACM 154
Branch-to-Branch 19
Brazilian 200, 210
Break Out 125, 215
set 107
Break Out Dialing 107
Breakout 152, 199
Reception 219
Breakout Dialing 152
BRI 25, 30, 32, 34, 37, 111, 140, 142, 294
BRI ISDN 140
BRI S-interfaces
ISDN 37
BRI So8 25
BRI-4 25
BRI-8 25, 27
Bridged Appearance 10, 109, 111, 112
bridged appearance button matches 112
Bridged Appearance Buttons 111, 112
Built-in IP 149
Business moving 282, 283
business needs 113
business requiring 180 283
business requiring 190 282
business-critical 161
business-to-business 162
Busy
DND 184
Busy Lamp Field 117, 152, 173, 180, 184

© 2008 AVAYA All rights reserved.

Busy Lamp Field Panel 184
Busy On Held 125
Busy Subscriber
Call Completion 142
Busy/Engaged 212
buy/lease 6
bypass
DND 106
Bytes 135

- C -

C3000 205
Cable 24, 27, 30, 40, 41, 42, 88, 97, 132, 164, 257, 291
Cable Lengths 291
Cable Modems 164
CALA 97, 277
Call 63, 115, 120, 158, 162, 168, 193, 198, 217, 252, 265
Advice of charge during 142
Call Activities 261
Call Appearance 86, 109, 111, 112
Call Appearance button 109, 111, 112
Call Appearance/Feature 46, 62, 85, 86, 87
call arrives 184
Call Back Sender 219
Call Back When Free 101, 102, 152
Call Barring 113, 114
call billing 194
Call Center 20, 84, 122, 193, 205, 213, 251, 252
Call Center View 251, 252
Call Center View provides
Supervisors 252
Call Center View Real Time Example 252
Call Completion
Busy Subscriber 142
call costing 113
Call Coverage 105, 111, 112
Call Coverage Buttons 111, 112
Call Data 10
Call Detail Records 265, 271
Call Details 265, 271, 273
Call Details Panel 184
Call Duration 176, 184
Call Flow Name 253
call flow programming interface 210
Call Flow Utilizing Database Actions 209
Call Flows 10, 209, 210, 211, 229, 253
Call Forwarding 18, 99, 115, 173, 211
Call Handling 6, 20, 99, 104, 173, 180, 184, 253
Call History 117, 173, 176, 180, 184, 190
Call Hold 151, 154
Call Identifier 253, 288
Call Intrude 106, 125
Call List 125
Call Listen 125, 200
Call Log 46, 58, 60, 64, 65, 66, 100, 107, 113, 176, 180
Call Park slots/zones 173
Call Pickup 108, 125
Call Pick-up 81, 152
Call Pickup Any 125
Call PickUp Extension 122
Call Pickup Extn 125
Call Pickup Group 125
Call Pickup Line 125
Call Pickup Members 125
Call Pickup User 125
Called/Calling Line ID Presentation 141
Called/Calling Name 141, 151
Called/Calling Name Presentation 141
Called/Calling Number 151
Caller Display 41, 89, 118
Campaign Manager 20, 200, 215, 219
Can Intrude 106, 287
Cancel All Forwarding 125
Cancel Ring Back When Free 125
Captaris 205
Carriage Return 271
carrying
Fax 136
SAP 161
CAS 32, 34
Cascaded Out-calling 214
Castelle 10, 205
Castelle Fax 10, 205
Castelle Fax Server 10
Castilian 200
CAT 132
CAT5 132, 291
Catalyst 132
CCBS 142
CCC 25, 107, 120, 152, 200, 251, 253, 257
CCC Reporter 251, 253
CCC System Administration 251
CCC Version
Microsoft CRM™ Reporting Integration New 253
CCM WB/22 257
CCP 295
CCV 251, 252
CCV Supervisory Screens 252
CD 217
CD-Rom 194
CDRs 113, 271
CE/UL/Dentori Safety Approved 291
cell/mobile 208
Central Office 30, 32, 34, 142, 161, 162, 163
Centralized 10, 22, 99, 140, 149, 152, 154, 197, 200, 218, 219
centralized billing 10

© 2008 AVAYA All rights reserved.

Centralized INTUITY Audix 197
Centralized Messaging 218
Centralized VM 218
Centralized Voice Mail 152
Centralized voicemail 10, 22, 140, 154, 200, 219
centralized VoiceMail Pro 10
Centralized Voicemail Services 219
Challenge Handshake Authentication Protocol 166, 295
Channel Associated Signaling 142
Channel Monitor 125
Channel Voice Compression 277
CHAP 163, 166, 168, 169, 295
Character Display 85, 87
Charger Unit 82
checkbox 206
Checking 142
Circuit ID
  Incoming 271
  Outgoing 271
Circuit Switched 141
Circuit Switched Data Call/Basic 154
Circuit Switched Data Call/Basic Call 141, 154
Circuit Switched Voice Networking 141
Circuit-switched 128
Cisco 84, 132
Cisco Systems 132
Citrix 181
C-LAN
  DEFINITY 9.5 154
Clear Call 125
Clear CW 125
Clear Hunt Group Night Service 125
Clear Hunt Group Out
  Service 125
Clear Hunt Group Out Of Service 125
Clear Quota 125
CLI 91, 180, 253, 286, 294
  existing 10
CLI Feature 91
CLI Schemes 294
CLI/ANI 200
Client PC 260
CLIP 142
CLIR 142
Clock 164
  Speaking 200, 219
closet/switch
  wiring 132
Co-Ax 277
Code Dialed 271
Code Used 113, 271
Codec G.711 69
Codecs 55, 56, 58, 60, 64, 65, 66, 134, 154
Cold Start 266
Collaboration 173, 176
Color Backlight Display 60
COLP
  Inhibits 142
COLR 142
Comfort 46, 72, 96
comma-separated 194
Common Features 55, 56, 58, 60, 62, 64, 65, 66
Compact Business Center 20, 22
Compact Call Center 251
Compact Contact Center 20, 200, 251, 253
Compact Contact Center Version 253
Compact DECT 10
Compact Flash 277
Compact Mode 176, 180
companies' LANs 128
company's
  IP Telephony unites 128
compared 69
  Service Provider conferencing 228
Compressing
  IP/UDP/RTP Headers 295
Compression Control Protocol 295
Computer Telephony Integration 24, 193, 194
Concise MIB Definitions 295
Concurrent Calls
  Maximum Number 219
Condition Code 271
Conference Add 125
Conference Bridge 6, 121, 229, 230, 231, 232
Conference Calling 18, 124
Conference Control Display 180
Conference entry/exit 231
Conference Held Calls 184
Conference ID 232, 237
Conference Meet Me 125
Conference Resources 230
Conference Room 97, 184, 189
conferencing 18, 24, 25, 39, 55, 56, 58, 60, 82, 85, 86, 87, 100, 117, 173, 184, 189, 228, 229, 230, 232, 273
  benefits relating 231
  Eliminating 6
  manage 231
  relating 231
Conferencing Center 22, 25, 124, 173, 180, 184, 230, 232, 237, 238
  adding 232
  IP500 232
  Requirements 238
Conferencing Center application 232, 237, 238
Conferencing Center Integration 238
Conferencing Center Reporting 232
Conferencing Center Scheduler 232, 238
Conferencing Center Server PC Specification 238

© 2008 AVAYA All rights reserved.

Conferencing Center toolbar 173
Conferencing Center Web 232, 237, 238
Conferencing Center Web Client 232, 237, 238
launch 238
Conferencing Center Web Client offers 237
Conferencing Center Web Scheduler 232
Conferencing Center Web Scheduler offers 232
conferencing service 6
Conferencing tone 100
Conferencing tone depending 100
congestion-control 137
Connected Line Identification Restriction 142
connection-oriented 149
Contact Activity 253
Contact Center 20, 24, 32, 46, 96, 122, 176, 209, 215, 251, 252, 253, 257
Contact Center Features 122
Contact Center Summary 253
Contact Management 20, 176, 251
Contact-ability 211
contactable 68
ContactStore 22, 25, 200, 216, 217, 223
ContactStore Search 200
Continuous Loop Greeting 219
Control Unit Conference Capabilities 230
controlled according 217
controlled-load 161
controlling calls 114
copier/scanner 205
CoS 136
Country Availability 281
CPE 165
CPU 217
CPU loading 217
CRC 142
Create' 213
CreateCall 286
CRM 32, 205, 253
CRM application
Small 253
Crystal 251
types 253
Crystal Design 253
Crystal Reports 251
purchase 253
Crystal Reports products 253
Crystal Reports Training 253
Crystal Training
World-Wide Source 253
Crystal/Business Objects 253
CS-ACELP 137
CSU 30, 34, 142, 270
CSU Loop-Back 270
CSU/DSU 30, 34, 142
CSV 251, 253
CTI 20, 25, 105, 193, 194, 195, 231
CTI interoperability
levels 193
CTI Link Lite 193
CTI Link Pro 25, 193, 194
CTI Link Pro RFA 194
CTI middleware 193
CTR3
ETSI T-Bus Interface 294
CTR4
ETSI T-Bus Interface 294
CU 253, 257
cust 104
Custom Reporting 251, 253
Customer Benefits 262
Customer Contact 241
Customer Contact Center 241
Customer Tracking 253
Customized Voicemail 219
CW 125
CW1308 291
Cyclic Redundancy 142
Czech Republic 281

- D -

D Message 288
Data Access Components 209
Data Call 119, 121, 141, 166
Data Channels 32, 134
Data Communication Solution 19
Data Communication Solution Features 19
Data Compression 166
Data Header Compression 166
Data Jack 90
Data Rates 294
data traffic 72, 135, 149, 165
Database 100, 200, 205, 209, 210, 215, 216, 217, 251, 253
Datagram Protocol 295
Date Records 271
Date Routing 10
Day One
starting 6
day/month 271
Day/Night service 94
D-channel 16kbps 294
D-channel 64kbps 294
DDI 10, 102, 142, 253, 286
DDI Call Duration 253
DDI Distribution 253
DDI Response 253
DDI Routing 253
DDI Summary 253
DDI/DID 10, 69, 99, 119, 142, 208, 252
DDI/DID call 119
DDI/DID Details 252

© 2008 AVAYA All rights reserved.

DECT 68, 69, 142
DECT Networking 69
Default
E&M 294
default greeting 103
default numbering 106
Definable PIN Code 219
DEFINITY 46, 85, 154
DEFINITY 6400 46
DEFINITY 9.5
C-LAN 154
DEFINITY G3si 154
DEFINITY/ACM
branches 154
DEFINITY/ACM occupying 154
delay-sensitive 161
Delete Message 219
deleting 103
Delphi 286
Delta Server 272
Denmark 281
Deploying
IP 132
Depth 80, 253, 291
Designing
IP Telephony 130
Reports Using 253
Reports Using Crystal Reports 253
Desk/Wall Mount 90
deskers 107
desking 10, 107
Desktop PC Telephony Controls 173
DevConnect 195
Developer Connection Program 195
Developer Edition 253
DeveloperConnect 195
Developers 193, 194, 195, 253, 287, 288
Development Solutions 253
DevLink 194, 288
DevLink Lite 194
DevLink Pro 194
DevLink Reserved Fields 288
DHCP 19, 75, 78, 136, 164, 168, 231, 295
DHCP Server 164, 168, 231
Dial 3K1 125
Dial 56K 125
Dial 64K 125
Dial calls 109, 113
Dial CW 125
Dial Direct 125
Dial Direct Hot Line 125
Dial Emergency 113, 125
Dial Extn 125
Dial Inclusion 125
Dial On Pickup 124
Dial Pad 184, 189
Dial Paging 125
Dial Plan 106
Dial Speech 125
Dial V110 125
Dial V120 125
Dial Video 125
dial/BLF 176
Dialed Number 100, 102, 114, 142, 176, 217, 271
controlling 117
Dialed Number Identification String 142
dial-in/dial-out to/from 163
Dialing In 32, 134, 142, 163, 168, 237
dialing plan 106, 140
DialPhysicalExtensionByNumber 125
DialPhysicalNumberByID 125
Dial-Up Circuit Support 164
DID 102, 142
DID/DDI 142, 176
Differentiated Services Field 295
Diffserve 149
Digital Station 16 Module V2 277
Digital Station 30 277
Digital Station 30 Module V2 277
Digital Station Expansion Module 40, 42, 85, 86, 87, 88
Digital Telephones 22, 71, 82
Digital Wireless 3810 Telephone 82
Digital Wireless North American 68
Dimensions 80, 291
Direct Dialing 99, 109, 208
Direct Dialing In 142
Direct Inward Dialing 102
Direct Media 63, 131, 134, 200
Direct Media Path 131
Direct Sequence Spread Spectrum 75
Direct Station Select 40, 42, 117, 173
Direct Station Select icon 173
Directory 6, 19, 85, 91, 100, 117, 118, 140, 168, 176, 184, 189, 287
Directory Entry 118, 184
Directory List 118
Directory Panel 184
Directory.csv 266
Disable ARS Form 125
Disable Internal Forward Busy 125
Disable Internal Forward Unconditional 125
Disable Internal Forwards 125
Disable Speakerphone 10
discover
NAT 146
Disk Space 217, 223, 253
Display Backlight 60, 76
Display Msg 125
Display Navigation Keys 86, 87

© 2008 AVAYA All rights reserved.

Display Soft Keys 86, 87
Distinctive 102, 176, 180
Distinctive Ringing 102, 176, 180
distinguish
  doorphone 94
distribute 120, 152, 176, 194, 205, 257
  voicemail 213
Distributed Hunt Groups 120, 152
Disturb 104, 195
DMS100 142
DMS-100 294
DMS-250 294
DMZ 146
DND 106, 115
  bypass 106
DNIS 142, 217
DNS 164, 166
Do Not Disturb 104, 106, 109, 115, 117, 125, 173, 195, 287
Do Not Disturb Exception Add 125
Do Not Disturb Exception Delete 125
Do Not Disturb Off 125
Do Not Disturb On 125
Domain 164, 166, 223, 237
Domain Name Service 164, 166
Door Entry 25, 32, 94, 108, 124, 176, 189
doorphone
  distinguish 94
doorphone calls 94
Doorphone Entry 94
Doorphones 94
DS 21, 27, 40, 42, 46, 56, 58, 60, 64, 65, 66, 82, 291, 295
DS 16 Module 291
DS 30 Module 291
DS Field 295
DS Phones 56, 58, 60
DS30 86, 87
DSS 40, 42, 64, 65, 66, 67, 86, 87, 97, 117, 215
DSS key 215
DSS key incorporates 215
DSS Modules 64, 65, 66, 67
DSS Unit 97
DSS/BLF 101
DSS/BLF key 101
DSS4450 86, 87, 88
DSS4450 provides 88
DSS4450 Unit 88
DSSS 75
DSU 30, 34, 142
DTE 168, 294
DTE Port 168, 294
DTMF 41, 90, 94, 114, 131, 137, 142, 157, 195, 208, 209, 215, 295
  sending 180
DTMF Dialing 90
DTMF Digits
  RTP Payload 295
DTMF signaling 41
DTMFA 294
DTMFC 294
DTMFD 294
D-Type 294
D-type on IP412 294
Dual Charger 72
Dual PRI E1 27
Dual PRI E1R2 RJ45 27
Dual PRI T1 27, 283
Dual Universal PRI 25
Duration Summary
  Incoming 253
Dutch 80, 81, 200, 210
DVD 217, 223
Dynamic 19, 55, 56, 58, 60, 64, 65, 66, 149, 161, 253, 260, 295
Dynamic Host Control Protocol 295
Dynamic IP 55, 56, 58, 60, 64, 65, 66
Dynamics CRM 260

- E -

E&M DID 142
E&M Switched 56K 142
E&M Tie Line 142
E.164 295
e.g 114, 157
E1 10, 21, 30, 32, 34, 111, 141, 142, 154, 218, 230
  including 18
E1 ISDN 230
E1 PRI 111
E1/PRI 277
E1/T1 154, 164
E1R2 10, 30, 32, 34, 111, 142, 277
E1R2 Channel Associated Signaling 142
E1R2 MFC 30, 34
E1R2 Primary 277
E911 124
Earth Loop Recall 41
Echo 34, 36, 134, 277
  128ms 18
e-commerce
  Internet 162
ECT 142
effectively—reducing 262
eg 104
Eliminating
  conferencing 6
ELR/TBR switchable 91
Email 6, 24, 72, 103, 162, 163, 167, 197, 198, 200, 223, 232, 251, 253, 270
  accessing 206
  Copy 219

© 2008 AVAYA All rights reserved.

Email 6, 24, 72, 103, 162, 163, 167, 197, 198, 200, 223, 232, 251, 253, 270
Forward 219
Host 237
leading 205
reading 20
sends 217
subject 202
Voicemail 202
Email addresses 206, 217, 232
Email application 202
email inboxes 20
Email Notification 202, 232, 270
Email Reading 206
email reading checkbox 206
Email Systems 20, 202
email WAV 198
emails 200, 202, 206
e-mails 149
Embedded 21
Embedded Applications 55, 56, 58, 60, 64, 65, 66
Embedded Messaging 21, 103
Embedded Messaging Card 270
Embedded Voice Memory 294
Embedded Voicemail 10, 20, 21, 25, 121, 125, 134, 197, 199, 219, 270, 277
Enable ARS Form 125
Enable Internal Forward Busy 125
Enable Internal Forward Unconditional 125
Enable Internal Forwards 125
enabled/disabled 106
Encapsulation 165
Frame Relay 137
encoding
G711 75, 78
Enlarged earpiece 79
Enterprise 72, 84, 181, 205, 228
ENTITY MIB 295
entry/double 231
Entry-level voicemail 199
EnumerateAddresses 286
equating
Exchange User 223
Equisys 205
ERP 205
escalates
doctor’s mobile/cell 214
ESP 295
ESP DES-CBC Cipher Algorithm 295
Estimated Time 258
Answer 219
Estonia 281
ETA 219
Ethernet 10, 19, 21, 25, 32, 37, 55, 56, 58, 60, 62, 64, 65, 66, 69, 72, 128, 132, 146, 161, 164, 277
Ethernet eliminates 132
Ethernet LAN 72, 128
Ethernet Ports 21, 55, 56, 58, 60, 64, 65, 66, 146, 161, 164
Ethernet Switch 10, 19, 161, 164, 277
EU24 40, 42, 60, 62
Euro ISDN 30
Euro-ISDN BRI 277
Excel 237
including 251
Except 63, 67, 88, 116, 146, 149, 230, 232, 270, 277
4601 132
Exchange/SMTP
GFI FAXmaker 205
executing
Acquire Call 122
Executive Wireless 97
Exit Queue 219
Expansion 10, 21, 24, 25, 27, 32, 37, 39, 40, 41, 42, 43, 55, 56, 58, 60, 62, 67, 86, 87, 142, 270, 273, 277, 282, 283, 291
Expansion Cards 277
Expansion Module Digital Station 30 277
Expansion Module Phone 30 277
expansion module provides 39, 43
Expansion Modules 21, 25, 32, 39, 40, 41, 42, 43, 60, 67, 86, 87, 270, 277, 291
Expansion Units 291
expiry 10
Explicit Call Transfer 142
Explicit IV 295
Exporting 266
Extended Callback Control Protocol 166
Extended CBCP 166
Extended Personal 200, 212, 219
Extended Personal Greetings 200, 212
Extension Activity 252
EXTENSION NUMBER 266
extension offering mobility 179
extension/VCM 277
External 37
External Bell 294
External Call 102, 106
External Control 124
External Control Port 124
External Expansion Modules 37, 277
External Number 115, 121, 142, 173, 199
External O/P 25, 32
External Participants 230
External Systems 176
Forward Emails 219
External Transferred Account Code 253
Extn Login 125
Extn Logout 125
extranet 237

© 2008 AVAYA All rights reserved.

Extreme Alpine Series 131
Extreme Alpine Series switches 131
Extreme Networks 131

- F -

failover 37
Fall Back 119, 162
Fast Forward 103, 199, 219
Fast Forward Message 219
Fast Start 131
Fax 10, 22, 39, 131, 137, 154, 173, 199, 205, 206, 219
  carrying 136
  routing 200
fax calls 131, 136
Fax Messages 205, 206
Fax Relay 137
Fax Transport 131
Faxination 205
FaxMail Pro 205
FCC 75, 78, 294
Feature Flag 271
Feature Key 22, 25, 32, 56, 69, 277
Feature Licensing 22
FEATURE NAME 266
Feature Support 219
Feature Table 90
Fenestrae 205
Fenestrae Faxination Server
  Microsoft Exchange 205
Field Data 288
FIFO 10
File Transfer Protocol
  including 167
Finland 281
Finnish 80, 81, 200
firewall 19, 24, 149, 162, 163, 167, 231
  Small Office Edition offers 164
firewall/VPN 6
firewalled 161
firewalled Layer 161
firewalls 167, 168
first-in/first-out 10
Firstly
  TTS 206
Fixed Feature Buttons 55, 56, 58, 60, 64, 65, 66
Fixed Feature Keys 85, 86, 87
Fixed Wallboards 251
Flash Hook 125
Flash Memory 199
Flow Control 56, 58, 60, 64, 65, 66, 137
Follow Me 109, 115, 116, 125, 200, 211, 287
Follow Me Here 125
For integrating 253
For report 253
Force login 287
Forced Account Code
  set 113
Forward All 117, 195
Forward All Calls 195
Forward Busy 115
Forward Emails
  External Systems 219
Forward Hunt Group 115, 125, 287
Forward Hunt Group Calls Off 125
Forward Hunt Group Calls On 125
Forward Message 219
Forward No Answer 115, 122
Forward Number 125, 287
Forward on Busy 112, 115, 116, 125, 195, 287
Forward On Busy Number 125, 287
Forward On Busy Off 125
Forward On Busy On 125
Forward on No Answer 104, 115, 125, 195, 287
Forward On No Answer Off 125
Forward On No Answer On 125
Forward Unconditional 115, 125, 287
Forward Unconditional Off 125
Forward Unconditional On 125
Forward voicemails 154
Forwarding 116, 184
  Email 219
  Multiple 219
  Multiple Mailboxes 219
  voicemail 200, 202
Forwarding except 116
FRAD 149
frame instructs 137
Frame Relay 134, 161, 163, 165, 200, 295
  Encapsulation 137
  framed 149
Frame Relay Assembler Disassembler
  IP Office employs 149
Frame Relay Encapsulation 295
Frame Relay's PVCs 149
framed 134, 137, 161, 163, 165, 200, 295
  Frame Relay 149
Front Door 94
FSK 294
FTP 167

- G -

G.711 55, 56, 58, 60, 64, 65, 66, 135, 137
  including 131
G.711 A-law/U-law 137
G.711A 154
G.711MU 154
G.723 69
G.723.1 135, 137
G.723.1 MP-MLQ 137
G.723.1. 131
G.723.1-6K3 154

© 2008 AVAYA All rights reserved.

G.726 16kbps ADPCM 217
G.729 69, 137
G.729 Annex 137
G.729a 78, 131, 135, 154
G.729a/b 55, 56, 58, 60, 64, 65, 66
G711
encoding 75, 78
Gatekeepers 128, 131
requests 130
Gateway 10, 18, 72, 84, 128, 130, 131, 134, 149, 161, 163
Greetings 176, 199, 200, 212, 219
Greetings & Mailbox Navigation 199
greetings provide 212
Ground Start 18, 34, 37, 39, 43, 142, 277, 294
Ground-Start 142
GROUP 266
Group Details 252
Group Listen 10, 125
Group Listen Off 125
Group Listen On 125
Group Monitor 252
Group Status 252
Group working 69
grouping incoming 161

- H -

Handover 69
Hands Free Pouch 72
Handset 10, 25, 64, 65, 66, 69, 71, 72, 80, 81, 82, 90, 99, 103, 108, 118, 120, 149, 165, 179, 193, 194, 200, 214
Handset Liquid Crystal Display 82
handset offers
high-resolution 72
Handset Volume Control 82, 90
Hands-free/ Speakerphone 81
Hardware 10, 24, 32, 69, 181, 190, 223, 238
Hardware provides 24
Hardware Support 10
Head Office 21, 118
Header Compression 136, 137, 295
Header Message 219
Headset 10, 55, 56, 58, 60, 64, 65, 66, 72, 80, 81, 82, 91, 96, 124, 125, 179, 181
Headset Socket 55, 56, 58, 60, 64, 65, 66
Headset Toggle 125
headset/microphone 128
Heat Dissipation 291
Held Calls Panel 184
Held Panel 184
Help Desks/Support Desks 209
HH 266
High Voltage 41, 91
Historical Reporting 251, 253
Hold 6, 55, 56, 58, 60, 82, 85, 86, 87, 90, 100, 101, 102, 106, 107, 117, 125, 161, 164, 173, 184, 212, 219, 258, 262, 272, 286
Hold Call 101, 125, 184
Hold Call Waiting 101
Hold CW 125
Hold Music 100, 101, 125, 258
Hong Kong 281
Hook 118, 124
Hook Dialing 118
Hot Desk 107, 152
Hot Desking 107
Hours greeting 121
HP's Network Node Manager 270
HTML 251
HTTP 167
Hunt Group Broadcast Messages 213
hunt group call handling 115
Hunt Group call ringing 108
Hunt Group Calls 108, 109, 115, 120, 121, 258
Hunt Group Disable 125
Hunt Group Enable 125
Hunt Group Enable/Disable 105
Hunt Group exceed 10, 123
HUNT GROUP EXTENSION 266
HUNT GROUP NAME 266
Hunt Group Recording 216
HuntGroup.csv 266
Hybrid" PBX 128
Hz 132

- I -

I55 154
Iceland 281
ICLID 142
Illuminated 3-line 80
Illuminated 5-line 81
IM 180
send 173
Immediate Reboot 266
Improved IP DECT Licensing 10
IMS 223
IMS Pro Connection 223
IMSAdmin 223
Inbound Call Handling 119
Inbound Call Operation 261
Incoming Call 10, 105, 118, 119, 142, 166, 173, 176, 180, 184, 197, 200, 216
Incoming Call Routes 10, 105, 119, 197, 200, 216
incoming call scripting 176, 180
IND CP CCM WALLBRD 257
IND DISP CCM WALLBRD 22 GB 257
India 281
indicating
Talk 82
Individual Agent Details 252

© 2008 AVAYA All rights reserved.

Individual DDI/DID Details 252
Individual Group Details 252
Individual Trunk Details 252
individual/team 253
industry-standard 154
Information Bulletin Boards 209
Information Protocol
Routing 168, 295
Information within Call Flows 209
Inhibits
COLP 142
Inline Power 10/100 BaseT Switching Module 132
input rating
power 291
In-Queue Announcements 219
Insert 100, 209, 217
Installation Guide 71, 82
Instant Messaging 173, 181
INT 277
Integral 10/100 Mbit Layer 164
Integral Static 19
Integral T3 46
Integral T3 IP 46
integrated conferencing 232
Integrated H.323 Gatekeeper 18
Integrated Management Suite 270
Integrated Messaging 20, 219
Integrated Messaging Pro 20, 22, 223
integrating incoming 260
integrating incoming calls 260
Intelligent 6, 119
Interaction 6, 209, 215, 251, 262
Voicemail 202
Interactive Voice Response 6, 20, 24, 209
building 200
Interchange 154
Internal Call 102, 109, 288
Internal Daughter Cards 36
Internal Directory 152
Internal Modem 21, 32, 36
Internal Modem Card 21, 36
Internal Twinning 109
Internal User 173
Internal/External 212
Internal/External greeting 212
Internet Access 19, 32, 134, 162, 165
Internet browsing 167
Internet Explorer 217, 232, 237, 238
Internet Explorer 6.0 238
Internet IP Security Domain
Interpolation 295
Internet Key Exchange 295
Internet Protocol 128
Security Architecture 295
Internet Protocol Control Protocol 295
Internet Protocol refers
type 128
Internet Security Association 295
Internet Service Provider 167
Internet Standards/Specification 137
Internet Telephony 6, 18, 146
Internet Telephony Service Providers 146
SIP trunking 18
internets 295
Internetworking 154
interoperability 151, 195
messaging 154
Interoperable 154
interoperate 168
networking 18
interoperation 168
Interpolation
Internet IP Security Domain 295
Interquartz 91, 94, 97
Interquartz Doorphone 94
Interquartz Gemini 91, 97
Interquartz Gemini 9281-AV 97
Interquartz Gemini Phones 91
Introduction 46, 232, 253, 260, 265
IP Telephony 128
Intrude 106, 107, 122, 194, 200, 287
INTUITY 154, 180, 198, 200, 205, 206, 207, 212, 213, 214, 218, 219
Intuity Audix 154, 219
Intuity Audix 4.4 154
Intuity Audix Emulation Features 219
Intuity AUDIX™ 154
Intuity Emulation 197
Intuity Feature 219
INTUITY Mode 198, 205, 206, 207, 214
Intuity TUI 219
Invited 184
IP Address Assignment 55, 56, 58, 60, 64, 65, 66
IP application
LAN 134
IP Authentication Header 295
IP DECT 3700 190
IP DECT Capacities 69
IP DECT IPO STARTER KIT 69
IP DECT System 69
IP DECT Telephone 69, 80, 81
IP DECT Wireless Handset 97
IP Encapsulation Security Payload 295
IP extension calling 282, 283
IP Extensions 21, 131, 282, 283
IP Hard Phone 128
IP Office Management Utilities 265
IP Office Manager 21, 25, 102, 106, 112, 117, 142, 167, 206, 212, 213, 216, 223, 231, 266, 271, 273
IP Office PC SoftPhone 181

© 2008 AVAYA All rights reserved.

IP Office Phone Manager application maintains 117
IP Office Phone Manager Lite 231
IP Office Phone Modules 41
IP Office Professional Edition 25, 198, 200, 219, 230, 232, 277
IP Office Professional Edition Upgrade 25
IP Office Small Community Networking 101, 102, 104, 106, 108, 117, 151
IP Office Standard Edition 25, 230, 282
IP Office stores
  list 113
IP Office Support Fax 136
IP Office Supports 136, 168
IP Office Time Profiles 10, 216
IP Office Voicemail Pro CD 223
IP Office Voicemail Pro Intuity Audix Emulation Features 219
IP Office VoIP 136
IP Office's 146
IP Office's Directory 100, 118
IP Office's list 113
IP Offices Transit Network Selection 142
IP Office's WAN 149
IP Packet 137, 149
IP Packet Flow Control 137
IP PBX 128
IP Phone Adapter 132
IP Phone Power Adapter 132
IP Phones 24, 25, 56, 58, 60, 118, 122, 130, 132
IP Power 132
IP Security Document Roadmap 295
IP Service 162
IP Softphone 179
IP Telephone Power Consumption 132
IP400 3rd PRTY IVR RFA 209
IP400 Access Point RFA 277
IP400 Analog 16 21
IP400 Analog Trunk 16 21, 39, 43, 282, 283
IP400 Analog Trunk 16 Module 39, 43
IP400 BRI Card 34
IP400 BRI-8 277
IP400 Digital Station 40, 42
IP400 Digital Station V2 Module 40, 42
IP400 Dual E1R2 34
IP400 Dual PRI E1 34, 277
IP400 Dual PRI T1 34, 277
IP400 E1R2 34
IP400 Internal Modem Card 36
IP400 Modem 12 277
IP400 Modems 277
IP400 Phone 41, 102, 282, 283
IP400 Phone Manager PC SoftPhone RFA 10 282, 283
IP412 10, 19, 21, 24, 25, 32, 34, 36, 37, 67, 124, 131, 136, 161, 164, 197, 219, 230, 277, 283, 291, 294
IP412 Control Units 32, 277, 283, 291
IP412 Maximum Heat Dissipation 291
IP412 Office A-Law Base Unit 277
IP412 Office Mu-Law Base Unit 277
IP412 PRI 60 E1 283
IP412 providing 277, 283
IP412 Supports Two 64-party Conference Banks 230
IP500 19, 20, 21, 22, 24, 27, 34, 36, 37, 39, 40, 67, 85, 86, 87, 88, 124, 131, 136, 142, 152, 161, 164, 197, 198, 200, 219, 230, 277, 282, 291, 294
  Conferencing Center 232
  default 25
  Embedded Voicemail 10
  existing 30
  feel 10
  require 10
  voicemail 199
IP500 Analog Phone 30
IP500 Analog Trunk 16 10, 37, 39
IP500 Analog Trunk 16 Expansion Module 37
IP500 Analog Trunk 16 Module 39
IP500 Analog Trunk Card 30
IP500 BRI 10, 30, 37, 39, 142
IP500 BRI So8 10, 37, 39
IP500 BRI So8 Expansion Module 37, 39
IP500 BRI So8 Module 39
IP500 BRI Trunk Card 30
IP500 Cards 27
IP500 Compact Flash voicemail 20
IP500 Control Unit 25, 27, 277, 282
IP500 Digital Station 30, 37, 40, 282
IP500 Digital Station Expansion Module 37
IP500 Digital Station Module 40
IP500 DS 16 10
IP500 DS8 Extension Card 85, 86, 87, 88
IP500 Expansion Modules 10, 25
IP500 Feature Key 282
IP500 Feature Key A-Law 282
IP500 Legacy Card Carriers 27, 30, 34, 36, 277, 282
IP500 Media Card Voice Compression Module 32 277
IP500 Media Card Voice Compression Module 64 277
IP500 Phone 10, 37
IP500 Phone 16 10
IP500 Phone Expansion Module 37
IP500 PRI 10, 30
IP-based WiFi 68
IPCP 295
IP-DECT 69
IP-Office Manager 10

© 2008 AVAYA All rights reserved.

IPSec 10, 19, 149, 169, 295
IPSec Tunneling 169
IPSec Virtual Private Network 10
IPSec VPN 149
IP-telephony 132
Ireland 281
ISAKMP 295
ISBN
asked 210
enters 210
ISDN 18, 30, 34, 39, 119, 134, 137, 141, 161, 163, 164, 173, 180, 271, 277, 294
BRI S-interfaces 37
following 142
incoming 142
outgoing 142
ISDN Basic 142
ISDN Basic Rate 142
ISDN BRI S-interface 277
ISDN DSS1 142
ISDN MSN 119
ISDN Ports 294
ISDN Primary 142, 277
ISDN Primary Rate 142, 277
ISDN T-Bus Basic Rate Interface 277
ISDN/PRI 119
ITAddress 286
ITU-T Recommendation E.164 295
IVR 6, 20, 200, 209, 219
providing 210

- J -

Japanese 200, 210
joined/left 232
Juniper 84

- K -

Kbps 152
LAN 136
Point 136
Kentrox 84
Key Management Protocol 295
Key System 128
Keyboard Actions 189
Keyboard Mapping 189
Kit List 282, 283

- L -

L2TP
Securing 295
Lamp Operation 111
Language 80, 81, 104, 118, 195, 197, 199, 200, 206, 208, 210, 266
language depending 197
Lanyard 72, 78
Last Number Redial 91, 125
last-in/first-out 10
Latin 200, 210
Latin American 200
Latvia 281
launch 273
Conferencing Center Web Client 238
Layer Two 295
Layer Two Tunneling Protocol 295
LCD 82, 91, 103
LCP 166, 295
LCR 157
Existing 114
LCS 173, 180, 181
LDAP 19, 118, 168
leading
email 205
Learning
Tree International 253
Leased Line 19, 141, 161, 162, 164, 165
types 164
Leased Line Support 164
Least Cost 113, 114, 121, 140
Existing 157
Least Cost Routes 113, 121
Least Cost Routing 114
Existing 157
LED 10, 64, 65, 66, 91, 132
Legacy Card Carrier 10, 25, 30
License Key 22, 169, 194, 199, 206, 209, 257, 282, 283
License.csv 266
LIFO 10
LIFO/FIFO 10, 219
Lightweight Directory Access Protocol 118
Link Control Protocol 166, 295
Linked Numbering 158
Linked Numbering Scheme 158
Listen Only 10, 237
Lite 20, 176, 180, 197
Lithium Ion Battery 78
Lithuania 281
LLB 30, 34, 142
Local Area Network 128, 131, 161, 164, 168, 195
Local End Echo Cancellation 25ms 137
Local Phone Directory 180
Local Telcos 142
Locating
SIP Servers 295
Longest Call Waiting 257
Longest Waiting 120, 189
Loop Disconnect 41
Loop Greeting 219
Loop Start 18, 34, 37, 39, 43, 142, 277
Loop start/Ground 294
Loop-Start 30, 142
Lord Of The Rings 210

© 2008 AVAYA All rights reserved.

Lost Call CLI 253
Lost Calls 253, 257
Lotus Notes 202, 205
Low-Speed Serial Links 295
LS 34
Luxembourg 281

- M -

MAC 161
Mail Box
Remote Access 219
Maintainers Network Manager 270
make/receive
headset/microphone 128
Malicious Call Identification 142
Manage Personal Distribution Lists 176
Manage voicemail 176, 218
Manage voicemails 176
Managed Frame Relay Network 149
Managed IP VPN 134, 149
Management Tools 21
Manager 5.1 266
Manager application 107
Manager provides 266
Manager Start-up Banner 10
Manager Warning Dialog 10
manager/secretary 112
Managment Information 295
Manufacturer's Liquid 76
Many Avaya 102, 111, 117
Many Simultaneous Calls Can 135
MAPI 20, 202, 223
Mbps 137, 149
Mbps LAN 149
MCID 125, 142
MCID Activate 125
MCU 130
MDAC 209
Media 21, 130, 134, 137, 161, 189, 194, 206, 253, 270
Media Service Provider 194
Medical 6, 71
medical issues 6
Medium Enterprise 253
Medium-Size Companies 6
Meet Me conferencing 228
Meeting 104
Meet-Me 25, 229, 230
Meet-Me conferencing 25
Meet-Me Conferencing on IP500 230
Meet-Me Conferencing Solution 229
Memory 10, 82, 91, 101, 134, 277, 294
Menu Bar 184
MERLIN MAGIX 46, 85
Message Announcements 219
message archiving 223
message handling
individuals 200
message informing 258
Message light 90
message recording 223
Message Storage Capacity 197
message taking 258
Message Waiting 22, 64, 65, 66, 82, 91, 102, 141, 213, 218
Message Waiting Indication 41, 91, 102, 154, 219
Message Waiting Indicator 55, 56, 58, 60, 85
message waiting lamps 102
Message Waiting Light 90, 154
Messages Button 58, 60
messages consume 223
Messaging Card 21, 270
Microsoft application 166
Microsoft CRM 253, 260, 262
Microsoft CRM Sales Reports 253
Microsoft CRM screen 262
Microsoft Live 173, 180, 181
Mid Span Power 132
Middle East 281
Mid-Span 132
Mid-Span Power Distribution Units 132
Miercom 6
milli-seconds 135
ML-PPP 165, 295
MM 266
mm/inches 291
Mobile Twinned Call Pickup 125
Mobile Twinning 22, 25, 125, 211
mobile/cell 109, 115, 173, 198, 200, 202, 214
Mobile/Cell Phone 109, 115, 173, 198, 200, 202
Mobility 10, 68, 69, 71, 72, 82, 99, 107, 179
Mobility Solutions 68
Modify' 213
Modify' existing lists 213
Modular 21, 24, 197, 218, 251
Modular Messaging 218
Modular Messaging Voicemail 197
Monitor 10, 30, 34, 107, 118, 122, 142, 176, 180, 189, 194, 195, 216, 217, 230, 251, 252, 257, 265, 269, 270, 273
Monitor Calls 122, 251
Monitoring - Call Center View 251
month/day 271
Most Avaya 117
Most Common Destination
Outgoing 253
mounting/desk 85, 86
MP 295
MPPC 295
MS-CRM 253
MSDE 217, 251

© 2008 AVAYA All rights reserved.

MSP 194
Mu-law 277
multicast 161
multicast filtering 161
Multiclass 137
Multi-Class Extension
Multi-Link PPP 295
Multiclass Extensions
Multilink PPP 137
Multi-Level Tree Structure 219
Multi-Link 134, 165, 295
Multilink PPP
Multiclass Extensions 137
multiple Auto Attendants 10
multiple incoming 111
Multiple Mailboxes
Forward 219
Multiple Subscriber Number 142
Multiple Time Entries 121
multipoint 130
point 142
Multipoint Connection Units 130
MultiVantage 154
Music-on-Hold 277
Mute 56, 58, 60, 80, 81, 90, 91, 117, 237
Mute All 237
MWI 41, 102

- N -

NA 42, 68
NAME 266
Calling 184
NAT 19, 149, 167, 295
discover 146
Netgear 84
Netherlands 281
Network Address Translation 19, 149, 167, 295
Network Address Translators 295
network addresses 19, 149, 167, 295
Network Alchemy 25
Network Assessment 136
network called
Public Switched 128
network comprising 168
Network Manager 270
Network Manager application 270
network numbering 158, 167
Network Numbering Schemes 158
network ready 6
Network Requirements 134
Network Time Protocol 295
network-critical 161
Networked Messaging 200, 207
Networked Messaging Solution 207
Networking Features 152
Networking maximizes 140
networks converging voice 72
networks including private 128
New message 102, 163, 206
new/old/saved 103
new/repeat/answered/unanswered 91
NI2 142
Night Service 119, 120, 121, 287
Night Service Fallback 120
Pass 121
Night Service Group 120
No Answer 109, 115, 116, 125, 184, 287
No Answer Interval 115
No Answer Time 109
Node Numbering 158
Node Numbering Scheme 158
North American Primary Rate Interface 142
North American T1 142
Norway 281
Norwegian 80, 200, 210
Not Disturb 106, 115, 117, 173
Not Disturb Exception Add 125
Not Disturb Exception Delete 125
Not Disturb Off 125
Not Disturb On 125
Notice 6
Now there's 6
NTP 295
NULL Encryption Algorithm 295
Number service 142, 211
Number/Incoming Trunk Access Code
Calling 271
number/name 219
number' 101
Numeric Keystrokes 189
Nylon Pouch 72

- O -

OAI 72
occurrence/instance 10
Octel 154
Of Hours 212
Off Hook Current 294
Off Hook Operation 124
Off Hook Station 125
Offer/Answer Model 295
offering
high-resolution 72
high-resolution backlight 72
Off-Hook Station 124
office heating 200
Office LAN
accessing 32
Office-quality speakerphone 78
offline 173
Oldest 122, 217
oldest ringing/waiting 122

© 2008 AVAYA All rights reserved.

oldest ringing/waiting call 122
On Demand Call Recording 200
Open Application Interface 72
Open CTI 20
Open Shortest Path First 161
OpenView application 270
Operator SoftConsole 20
Opportunity Activity 253
Optional Add-Ons 64, 65, 66
Other Avaya Products 46, 85, 154
Other Features 10, 219, 230
Other Ranges
  Telephones Compatible 85
Out Of Hours 212
Outbound Call Handling Features 113
Outbound Call Operation 262
outcall 214
Outcalling 219
Outlook 20, 176, 180, 205
  Goldmine 180
Output Port 94, 294
Overflow Group 120
Overhead LAN 135
Overhead WAN 135

- P -

PABX's 18, 130
  provide 21
Packet Based 149
Packet Based Voice Networking 149
packet filtering 167
packet switching 149
packetization 137
packetized VoIP
  allowing 149
packetized VoIP calls 149
packets arrive
  port 137
packet-switched 128
Packet-Switched Telephony 128
Packet-switched VoIP 128
Pager 202
Paging
  Calls 109
  receiving 41
Pakistan 281
Pan European Connection 294
PAP 163, 166, 168, 169, 295
Park 101, 102, 109, 125, 173, 184, 189, 190, 200, 215, 288
Park Call 125
Park ID's 184
Park Return 109
Park Slot Panel 184, 189, 190
Park Slots 101, 184, 189, 190, 215, 288
pass calls 121
Password 10, 103, 163, 166, 190, 232, 266, 295
PC application 118, 213
PC application Phone Manager Pro 213
PC Based 197
PC CTI application 251
PC Requirements 223
PC Softphone 20, 25, 99, 179, 180, 181
PC Wallboard Example 257
PC-based 251, 253
PCMCIA 277
PCMCIA Slots 277
PCMCIA Wireless 277
Percentage Time 252
Permanent Virtual Circuits 149, 165
Personal 211
Personal Distribution Lists 176, 180, 213
Personal Numbering 200, 211, 219
Personal Options 219
Personal Productivity 173, 176
personalization 176
Personalized Greeting 219
Personalized Ring Patterns 55, 56, 58, 60, 64, 65, 66
Personalized Ringing 90, 102
Phone 16 277, 291
Phone 16 Module 277, 291
Phone 16 Module V2 277
Phone 30 Module 277, 291
Phone 30 Module V2 277
Phone Call Activities 261
Phone Manager application 105, 106
Phone Manager Conferencing Center Integration 238
Phone Manager Feature Summary 180
Phone Manager GUI 212
Phone Manager Lite 20, 25, 117, 172, 173, 176
Phone Manager Lite/Pro/PC Softphone 25
Phone Manager PC Softphone 22, 128, 172, 179, 181, 251
Phone Manager Pro 10, 20, 22, 25, 94, 108, 117, 172, 176, 179, 180, 181, 212, 213, 219, 251, 282, 283
Phone Manager Pro PC Softphone 25, 282, 283
Phone Manger Pro 251
Phone V2 41
physical/logical 136
Pickup 108
Pick-Up 104
Pilot 253
Pilot Call Duration 253
Pilot Distribution 253
Pilot Response 253
Pilot Routing 253
Pilot Summary
  Incoming 253
PIN 20, 114, 198, 209, 219, 231, 232, 237

© 2008 AVAYA All rights reserved.

PIN 20, 114, 198, 209, 219, 231, 232, 237
prompted 197
requesting 229
PIN checking 209, 232
PIN Code By-Pass 219
PIN code/menu 231
PIN Restricted Calling 114
Plain Old Telephone Services" 128
Plain Ordinary Telephone 277
Platform Support 219
Play Advice
switching 216
played back 10, 212, 219
PoE 46, 55, 56, 58, 60, 62, 64, 65, 66, 69, 128, 132
Point Compression
Microsoft Point 166, 295
Point Protocol
Point 295
Point WAN 136
point-to-multipoint 39
Point-to-Point 39, 142, 149, 161, 163, 165
Point-to-Point Protocol 161, 165
Power Distribution Units 132
Power Fail Ports 294
Power Options
IP Telephones 132
Power Supply 55, 56, 58, 60, 62, 64, 65, 66, 88, 132, 277, 291
Power Supply Units 55, 88, 291
Powered Data Unit 132
Powered LAN 46
PowerPoint
reviewing 237
PPP 137, 161, 165, 169, 295
PPP Fragmentation 137
PPP MP 295
PPP Multilink Protocol 295
PR 142
Pre-licensed IP-DECT 10
PRI 10, 18, 19, 25, 27, 30, 32, 34, 140, 142, 282, 283, 294
PRI 30 E1 27
PRI 30 E1R2 RJ45 27
PRI 48 T1 282, 283
PRI 60 E1 282, 283
PRI E1 294
PRI ISDN Services 294
PRI ISDN Switch 294
PRI T1 19, 27, 283, 294
PRI T1 Service 294
PRI T1/J1 294
prices—Avaya 6
Primary Rate ISDN 142
Primary Rate Trunks 30, 34, 142
prioritization 72, 161
Priority 6, 56, 58, 60, 72, 119, 125, 128, 161, 214, 287
Priority Call 119, 125
Priority Processors 72
Privacy Mechanism
Session Initiation Protocol 295
Private 6, 30, 34, 107, 113, 125, 128, 140, 141, 149, 154, 162, 167, 213, 214, 237, 277
Private Call 107, 125
Private Call Off 125
Private Call On 125
Private Circuit Switched 141
Private Circuit Switched Voice Networking 141
Private Networking 154
Private Voice Networks 30, 34, 140
Pro 10, 20, 94, 202, 223, 231, 258
upgrades 22
Pro provides 20
Product 6, 10, 34, 71, 84, 85, 91, 122, 125, 154, 181, 190, 193, 194, 195, 205, 223, 238, 253, 277
receiving 6
time during 22
product configuration documents 125
Product Configurations 125, 277
Product Description 34, 122, 238
Operating System 181, 190, 223
Product Documentation 71
Professional Edition 10, 22, 230, 253, 282
Standard Edition 25
upgrade 25
prompting
PIN 197
Voicemail Pro 208
Protocol Applicability Statement 295
protocol passing 167
Protocols 30, 34, 69, 72, 128, 130, 131, 134, 137, 149, 154, 165, 167, 168, 205, 269, 295
protocols including 167
Proxy Address Resolution Protocol 167
PSK 78
PSTN 111, 134, 136
IP Office connects 128
SCN 114, 157
PSU 291
PTT 79
Public 6, 10, 19, 30, 34, 84, 102, 113, 114, 121, 128, 130, 136, 140, 142, 149, 164, 167, 169, 200, 213, 237, 295
Public Network 113, 114, 130, 149
Public Switched 136
network called 128
Public Switched Telephone Network 136
called 128
Public Voice Networking 142
Pulse 41, 108, 124, 142
Pulsed High Voltage 41

© 2008 AVAYA All rights reserved.

Push 94
push-to-talk" 72
Put_EventFilter 286
PVCs 149, 165

- Q -

Q.931 137, 142, 149, 151
Q.931 signaling 142
Q2 2007 Maintenance Release 10
Q3 2007 Maintenance Release 10
QoS 18, 55, 56, 58, 60, 64, 65, 66, 72, 134, 136, 149, 181, 273
QoS Options 55, 56, 58, 60, 64, 65, 66
QoS/Class
  Service 136
QSIG 18, 111, 142, 152, 154
  following 141
  limit 25
  running 218
  terminates 141
QSIG Networking 18
QSIG provides
  level 141
QSIG signaling 141
Quad Chargers
  3641 72
Quality Assurance 108
Questions & Voting 237
Queue Announcements 10, 197, 200, 219
Queue Based Screens 252
Queue Entry Announcement 219
Queue Handling 258
Queue Manager 20
Queue Mode 189
Queue Monitor 252
Queue Panel 184
queue panel displays
  bar 184
queue position 10, 219, 258
Queue Position Announcement 219
Queue Threshold Alert 10, 123
Queue Update Announcement 219
QUEUING 266
Quick Charger 72
Quotas 162, 163, 165, 288
Quotas place 162

- R -

Radio Frequency 2.4000 75, 78
Radio Frequency 2.4000 GHz 78
RAID 223
RAS 32, 134, 168
Rate Adaptation 295
Rating
  24V DC 291
reachability 6
reading
  email 20
Real Time 20, 137, 200, 251, 252, 295
Real Time Control Protocol 137, 295
real time prompting
  supervisor 251
Real Time Reporting 252
Real Time Status 252
Real Time Supervisor Monitoring 251
real time tracking 20
Reattempt 184
Reboot When Free 266
Rec 295
Recall button 90, 91
receiver's 128
Receiving Door 94
Reception 102, 124
  Breakout 219
Rechargeable Battery 82
Reclaim Call 105
recommendations 161
Record Message 125, 215
Record/Send 219
recorded message stating 215
Recording 216
  Time 219
Recordings 10, 215, 216, 217, 223
Redial Button 82, 90
Region 68, 69, 89, 97, 282, 283
RegisterCallNotifications 286
relating
  conferencing 231
Relay Off 125
Relay On 125, 200
Relay On/Off/Pulse 108
Relay Pulse 125
Release 4.0 273
Release 4.0.7. 69
Release 4.1 10, 121, 164
Release 4.1. 10
Remote Access 19, 21, 32, 36, 99, 134, 162, 163, 168, 179, 200, 277
  Mail Box 219
Remote Access Features 163
Remote Access Server 19, 21, 168
Remote Access Services 32, 99, 162, 163
Remote Hot Desking 107, 152
Remote Management 231
REN 41, 294
Repeat Message 219
Replay 200, 217
replay rights 217
Reply 199, 206, 212, 219
Report Design Solutions 253
Report Manager 253
Report Scheduler 251, 253

© 2008 AVAYA All rights reserved.

Report Viewers 251
Reporting Integration New 253
Reports Training 253
Reports Using
Designing 253
Reports Using Crystal Reports
Designing 253
requesting
Gatekeeper 130
PIN 229
reseller 253
Rest 161
World 230
restricted/allowed 167
Resume Call 125
Retrieve Call 125
Return On Investment 228
reviewing
PowerPoint 237
Rewind Message 219
RFA 154, 257
RFC 137, 295
Ringer 10, 64, 65, 66, 82, 90, 91, 102, 123
Ringer Equivalency 90
Ringer On/Off 82
Ringing
Patterns 90
Volume Control 90
RIP 19, 168, 295
RJ45 Ethernet 294
RMS 294
ROI 228
ROTARY 266
router alleviates 161
router/firewall/DHCP 21
Routing Information Protocol 161, 168, 295
RPT 253
RTF 253
RTP 135, 146, 295
RTP Payload
DTMF Digits 295
RTP Relay 146
RTP Voice Data Payload 135
RTP/RTCP 137, 295
Ruggedized 77, 78, 79, 97
Russia 281
Russian 200, 210
RW 217

- S -

S Message 288
S0 142
S0 Endpoint
Call 142
S3210 154
Sales Reports 253
SAP
carrying 161
Save Message 219
Save Profile 176, 184
SBC 146
S-Bus 39
scalable 21, 140, 205, 260
Scalable Platform 21
Scan' 213
Scheduler 232
SCN 25, 107, 120, 152, 173
PSTN 114, 157
Screen Pop 180, 194, 260
Screen-Popping 100
SDP 295
Search 189, 217
Secondary Dial Tone 106, 125
secretary's 106
Securing
L2TP 295
Security Architecture
Internet Protocol 295
Self-Administration 118
Send Email 219, 270
Send Instant Messages 173, 180
Separated incoming/outgoing 180
serialy 257
Series 10, 22, 27, 37, 40, 42, 46, 56, 58, 60, 62, 90, 97, 190, 195, 200, 291
Server - Base System 251
Server PC 223, 260
server PC's 291
servers provide 166
Service Packs 181, 190, 223, 238
Service Provider conferencing
compared 228
Service Provider conferencing services 228
service provider supporting 100
Service Providers 18, 24, 100, 142, 146, 149, 169, 173, 180, 228
Service Reports 253
service-by-service 165
Session Border Controller 146
Session Description Protocol 295
Session Initiation Protocol 128, 137, 146
Privacy Mechanism 295
Set Absent Text 125
Set Account Code 125
Set Authorization Code 125
Set Hunt Group Night Service 125
Set Hunt Group Out
Service 125
Set Hunt Group Out Of Service 125
Set Inside Call Seq 125
Set Message Priority 219

© 2008 AVAYA All rights reserved.

Set Mobile Twinning Number 125
Set Mobile Twinning Off 125
Set Mobile Twinning On 125
Set No Answer Time 125
Set Outside Call Seq 125
Set Ringback Seq 125
Set Wrap Up Time 125
SetCallInfoBuffer 286
short code representing 215
Short Codes 10, 101, 121, 122, 124, 125, 152, 165, 199, 215, 231, 257
Short Message System 202
shortcode 41
ShortCode.csv 266
Shut Down Embedded Voicemail 125
ShutdownEmbeddedVoicemail" 10
Signaling
  Conforms 142
Signaling Channels 69, 142
Silence Suppression 131, 137
Simple Network Management Protocol 270, 295
Simple Outlook 180
Simple Telephony Call/Basic 141, 154
Simple Telephony Call/Basic Call 141, 154
Simple Traversal
  UDP 146
  User Datagram Protocol 295
Simultaneous 136
Simultaneous VoIP Calls
  Maximum Number 136
Small Community Network 10, 101, 102, 104, 106, 107, 108, 117, 120, 151, 152, 154, 173, 200
SMDR 25, 113, 194, 271
SMI 75
SMS 202
SMS/text 205
SMTP 20, 200, 202, 223, 265
SMTP email 265
SMTP/POP3 205
SNMP 55, 56, 58, 60, 64, 65, 66, 132, 149, 265, 270, 295
SNMPv1 295
So8 32, 277, 291
So8 Module 32, 277, 291
Soft 18, 32, 78, 86, 87, 105
Soft Keys 86, 87
SoftConsole 22, 25, 101, 104, 108, 117, 124, 184, 189, 190, 232
SoftConsole Administration 190
SoftConsole gives
  operator 184
SoftConsole Options 189
SoftConsole PC Requirements 190
SoftConsole Telephone Requirements 190
Software Developers Kit 193
Software Development Kit 20, 194
Software License Feature Key A-Law 277
Software License Feature Key Mu-Law 277
software/documentation CD 277
SOHO 106
Sophisticated Queue Announcement 200
SOS Emergency key 80
soundcard 179
Speaker button 118
  pressing 10
Speaking Clock 200, 219
Special Features
  4625 SW 60
  5410 58
  5420 60
  5621 SW 60
Special Services 142
Specialty Handset Support 90
Spectrum 257, 260
Speed Dial 58, 60, 67, 117, 118, 173, 180
Speed Dial List 58, 60
Speed Dial/BLF 176
Speed-Dial/BLF 173
speed-dial/Busy Lamp Field 176
spread-spectrum 71
Sprint 142
SQL 209, 251, 253
SQL Query Builder Wizard 209
SSA 10, 273
SSA connects 273
SS-CNIP 141
SS-CNIR 141
SS-CONP 141
SS-CT 141
SS-MWI 141
SSS 142
STAC 295
Stac Lemple Ziv 166
STAC LZS Compression Protocol 295
stackable 25, 32
Stafford Technology 253
Stand 85, 86
  Charging 82
Stand Power Supply Adapter
  Charging 82
Standard 40W Power Supply Unit 291
Standard Edition 10, 22, 253
  Professional Edition 25
Standard Reports List 253
standards-based 18, 19, 119
standards-based TAPI 119
Start Call 176
Starter Kits" 69
starting
  Day One 6
  Manager 10

© 2008 AVAYA All rights reserved.

starting
PC Wallboard 257
State/Province 253
station/access 46
Status 10, 21, 25, 62, 75, 104, 107, 112, 117, 120, 173, 184, 189, 195, 200, 217, 237, 251, 252, 253, 265, 273
Status Application 10, 21, 25, 273
Status Bar 184
STD15 295
STD16 295
STD17 295
STD56 295
STD57 295
Still Queued 200
Stop Call 176
strings 142
Absence Text 104
Structured Query Language 209
STUN 146, 295
Sub-addressing
Allows 142
subject 6, 100, 122, 172, 216
email 202
subnet Phone Manager 181
Summary 30, 97, 136, 232, 237, 253
Positioning 197
Supervised Transfer
called 102
supervisor's Coverage Timer 109
Supervisors 105, 109, 213, 253, 257, 258
Call Center View provides 252
real time prompting 251
Supplementary Service 149, 151
Supplementary Services within 151
Supplementary Services within IP Networks 151
surfing
Internet 32
Suspend Call 125
Suspend CW 125
Sv 288
SW 56, 58, 60
SwapHold 286
Sweden 281
Swedish 80, 81, 200
switchable 58, 60, 91
Switchable Time Break Recall 100 91
Switched Ethernet 19, 25, 32
Switched LAN 21
switching
Capacity 294
Play Advice 216
WAN 149
synchronization 168, 219
Phone Manager 19
Syslog 10
System Administration 217
System Administrator 100, 102, 105, 125, 173, 216, 217, 232, 238
System Announcement 219
systems supporting 207

- T -

T.38 131, 136
T.38 Fax 136
T1 10, 18, 19, 21, 30, 32, 34, 102, 111, 114, 140, 141, 142, 154, 157, 164, 218, 230, 277, 282, 283
T3 10, 40, 42, 46, 63, 64, 65, 66, 67, 97, 103, 142, 291
Tabs 176, 180, 184, 189, 231, 232, 271
TAPI 20, 25, 100, 194, 195, 251, 286, 287
TAPILink Lite 194, 195, 286
TAPILink Pro 194, 195
TAPI-WAV 194, 195
Target Graphical Summary 253
Target Member Duration 253
Target RAS 288
TCP 295
TCP/IP 21, 223, 272, 295
TCP/UDP/IP 137
Technical Bulletin 146, 181, 190, 223, 238
Technical Specifications 181, 190, 223, 238, 272
Technology Overview 130
TEIs 39
telecommunication 295
telecommunication numbering 295
Telecommuter 10, 108, 176, 180
Telecommuter Mode 10, 108, 176, 180
teleconferences 6
Telephone Adaptors 21, 163
Telephone Cord 82
Telephone Devices 71, 195
telephone establish 130
Telephone Extension Cable Lengths 291
Telephone Options 22
Telephone User Interface 200, 212, 213, 214
telephone wishing 130
Telephones Compatible
Other Ranges 85
telephones including 18
telephones operating 137
Telephones Section 40, 42
telephones utilizing Power 132
Telephony Functions 99
Telephony Signals 295
Telephony Tones 295
telesales 200
Terminal Support 10
Test Conditions 219
Text To Speech 20, 22, 200, 206, 210, 219, 223
Text" 208

© 2008 AVAYA All rights reserved.

Text-to-Speech 200
TFTP 72, 75, 78, 136, 295
that's 6
that's right 6
There's 6
These Lite 22
they're 173
Third Party Database Access 219
Third Party Fax 205
Third Party Text
Speech 22
Third Party Text To Speech 22
Through Network Address Translators 295
TIA/EIA-646-B
conform 142
til 104
Time Division Multiplexing 24
time during
product 22
Time Entries 121
time linking
office 161
Time Profile 10, 107, 120, 121, 165, 214, 216, 287
timebands 163
Timed Break Recall 41
Time-Division Multiplexed Telephony 128
timeframe 219
timeout 10, 100, 101, 214, 287
timeout causes
user 10
TLS 10
TNS 142
To Email 200, 202, 219
Toggle Calls 100, 125
toolkit 194
toolset 20
toolset including 20
Topic 253
Total base-stations/repeaters 69
TPAD 168
Traditional Wall Mounted Wallboards 257
Transaction Packet Assembler Dissembler 168
Transfer call 10, 184, 253
operator 219
Transfer Call Tracking Detail 253
Transmission Control Protocol 295
transmission/reception 142
Transmit Power 100mw 75
Transport Layer Security 10
TransTalk 9040 10, 46
Tray 173, 184, 190
Tray working 190
Tree International
Learning 253
tri-color 257
trigger/control 124
Trivial File Transfer Protocol 295
Trunk Access Code
Incoming 271
Trunk Cards 27, 30, 34, 282, 283
Trunk Details 252
Trunk Group Activity 253
Trunk Group Busy 253
Trunk Group Call Duration 253
Trunk Group Details 252
Trunk Group Monitor 252
Trunk Group Response 253
Trunk Group Summary 253
Trunk Interface Cards 32, 34, 277
Trunk Interfaces 18, 21, 32, 34, 277
trunk lines 111, 282, 283
trunk providing 30
Trunk Related Screens 252
trunk/extension 30
trunk/VoIP 230
trunkinterfacecards.htm 34
Trunks back 21
trunks/VoIP 231
TTS 206
adds 210
TTS Licensing 210
TTY 200
TTY hearing 200
TUI 198, 200, 212, 213, 214
Tunneling Protocol 169, 295
Twinning
Appearance Keys 10
two-base-station 69

- U -

UK 200, 210
UK20 294
unencrypted 166
Uni 27, 277
Unique computer-generated 232
Unique computer-generated Conference ID 232
Unique PIN 232, 237
unique PIN numbers 232, 237
United Kingdom 281
Universal PRI 10, 30
Unlike circuit-switched 128
unmaps 167
Un-Mute All 237
Unpark 125, 286
Unpark Call 125
Unread 'User 184
unsecured' 164
Unshielded Twisted-Pair 291
Unstructured Private Circuit 149
Upgrade License 25, 230
upgradeable 55, 56, 58, 60, 64, 65, 66, 72

© 2008 AVAYA All rights reserved.

Upgradeable Firmware 55, 56, 58, 60, 64, 65, 66
Upgrade-Kit" 69
upgrade-license 69
Upn 63, 97, 291
UPS 132, 291
URL 232
US T1 PRI 277
USA 281, 294
USA Interfaces 294
USB 168, 179
use depending 30
Use mailing 219
user according
type 190
user acquiring
call 122
User Agents 71
User CD-Rom 194
user chooses 198
user collecting 163
User Datagram Protocol
Simple Traversal 295
user determines 112
user ensuring 97
user executing 122
User Interface 20, 78, 128, 198, 200
user interface offering 128
user making 116
user name 10, 184, 288
User Recording 216
user restricting 168
USER RIGHT 266
User Rights 114, 176, 266
Using Text To Speech 20, 210
UTP 291

- V -

V.24 164, 277, 294
V.24 Interface
19.2Kbps 294
V.24/V.28 294
V.24/V.35/X.21 277
V.32 30, 163
V.35 149, 161, 164, 277, 294
V.90 21, 32, 36, 163, 277
V.90 56Kbps 163
V110 295
V120 295
V24 164
V24/V35/X21 277
V35 164
V5.0 217
V90 19
VAC 132
Variable Routing' 10
VB 22, 200, 210
VB Scripts 22, 200, 210
VC 277
VCM 18, 25, 27, 32, 36, 63, 69, 108, 134, 136, 146, 149, 152, 219, 270, 273, 282, 283
VCN 232
Virtual 84, 128, 149, 165, 228
Voicemail 219
Virtual Office 84
Visual Basic 200, 210, 219, 286
Visual Basic Scripts 219
Visual Message Waiting Indication 91
Visual Voice 103, 199, 200, 219
Visual Voice NOT 103
VLAN 55, 56, 58, 60, 64, 65, 66
VM 115, 253
VM Call Flow Monitor 253
VM Summary 253
Voice Call 119, 135, 154
Voice Communication Solution 18
Voice Communication Solution Features 18
Voice Compression 21, 27, 32, 36, 69, 130, 131, 134, 136, 146, 149, 199, 277, 283
Voice Compression Module 32, 36, 130, 131, 134, 136, 149, 277
Voice Conferencing Notification 232
Voice encoding 78, 277
Voice encoding G711 75
Voice Forms/Questionnaire Mailboxes 200
Voice Mail 6, 80, 81, 115, 184, 214
Voice Messaging 18
Voice Networking 22, 25
Voice Networking Licenses 10
Voice Priority Processors 72
voice processing 194
Voice Recording 108, 200, 216
Voice Recording Library 223
Voice Recording Library Management 223
Voice Recording Library" 217
voice samples
telephone 134
voice traffic 72, 134, 136, 163
VoiceDirector 142
Voicemail 10, 20, 21, 22, 25, 32, 58, 60, 94, 99, 102, 103, 105, 106, 107, 108, 109, 112, 114, 115, 121, 122, 124, 125, 131, 134, 140, 152, 154, 163, 173, 176, 180, 184, 197, 205, 206, 207, 208, 209, 210, 211, 213, 214, 215, 216, 217, 218, 223, 229, 231, 232, 258, 266, 270, 273, 277, 287, 288, 291
alter 120
comment 198
control 199, 200
distribute 213
Email 202
Forwarding 200, 202
Hunt Groups 219
Individual 219

© 2008 AVAYA All rights reserved.

Voicemail 10, 20, 21, 22, 25, 32, 58, 60, 94, 99, 102, 103, 105, 106, 107, 108, 109, 112, 114, 115, 121, 122, 124, 125, 131, 134, 140, 152, 154, 163, 173, 176, 180, 184, 197, 205, 206, 207, 208, 209, 210, 211, 213, 214, 215, 216, 217, 218, 223, 229, 231, 232, 258, 266, 270, 273, 277, 287, 288, 291
Interaction 202
IP500 199
listen 198
PC 199
routed 212
Virtual 219
Voicemail Access 173
Voicemail application 198
Voicemail Box Feature 219
Voicemail Breakout/Personal Auto-Attendant 199
Voicemail Collect 125, 199
Voicemail email 211, 223, 287
Voicemail Email Connection 223
Voicemail email forwarding 211
Voicemail Email Integration 223
Voicemail Feature Comparison 219
voicemail greeting 20
Voicemail Help TUI 219
VoiceMail Lite 20, 25, 103, 121, 197, 198, 200, 202, 219
voicemail needs 198
Voicemail Node 125
Voicemail Off 125
Voicemail On 125, 198, 211, 287
Voicemail PC 223
Voicemail Ports 108, 134, 216
VoiceMail Pro 10, 20, 22, 25, 103, 108, 121, 124, 152, 154, 176, 184, 197, 200, 205, 206, 207, 209, 210, 211, 212, 213, 214, 215, 216, 217, 219, 229, 231, 232, 258, 270
406 32
500MB 223
License 223
pool 210
prompting 208
VoiceMail Pro application 94, 258
Voicemail Pro Client 200, 208, 209, 214, 216
VoiceMail Pro Fax 205
VoiceMail Pro Intuity 176
VoiceMail Pro Manager 208, 219
VoiceMail Pro Networked Messaging 22, 154, 213
VoiceMail Pro Networked Messaging RFA 154
VoiceMail Pro Server 154, 200, 206, 207, 210, 223
voicemail ringback 102, 125, 219
Voicemail Ringback Off 125
Voicemail Ringback On 125
Voicemail Server 154, 163, 197, 200, 223
Voicemail System 214, 217
Voicemails 10, 103, 108, 173, 176, 198, 202, 206, 219
VoIP 6, 18, 21, 22, 27, 36, 68, 72, 96, 128, 131, 134, 135, 136, 137, 140, 149, 154, 180, 273, 277
VoIP application 136
VoIP calls 135, 136
VoIP provides 128
VoIP Standards Supported 137
VoIP Wi-Fi Solution 68
VoIP-compatible 128
VoIP-ready VPN 10
Voltage 91, 132, 294
Volts 132
Volts Alternating Current 132
Volume 55, 56, 58, 60, 80, 81, 85, 86, 87, 90, 91, 117, 208, 219
Volume Control
Ringing 90
Volume Down 55, 56, 58, 60
Volume Up 55, 56, 58, 60, 85, 87
Volume Up/Down 86
VPIM 219
VPN 19, 21, 22, 25, 84, 149, 169
following 68
terminate 10
VPN IPSec/L2TP 22
VPN Phone Software 84
VPN Phones 10, 22, 25, 68
Licenses 84
VPN tunneling 21
VPN-access 84
VPN-gateways 84
VRL 223

- W -

waiting
Acquire Call 122
walkie 72
walkie-talkie 77
Wall Mounted Wallboards 257
Wall Plate Adapter 82
Wallboard Manager 257
Wallboard Manager Communications 257
Wallboard Manager/Wallboard Server 257
Wallboard Server 257
Wallboard Server/Client 257
Wallboard/22 257
WAN 10, 19, 21, 24, 69, 134, 135, 140, 161, 163, 164, 165, 169, 200, 269, 277, 294
switching 149
VCM 134
WAN Expansion Kit 277
WAN link 161
WAN multiplexers 163
WAN3 25, 37, 270, 277, 291
WAN3 10/100 25, 37, 277
WAN3 10/100 Module 277
WAN3 Module 270, 291

© 2008 AVAYA All rights reserved.

WAP WML 58, 60
Warm Start 266
Watts 132, 291
WAV 101, 176, 180, 189, 202, 206, 217
wav file 101, 180, 217
waveform 217
Web 6, 19, 71, 173, 215, 216, 217, 223, 232, 253
Logged 237
web address 232
Web Chat 237
Web Client 232, 237
Web Client offers 237
Web Scheduler 232, 237
Web Server 217, 223, 237
Web Server Operation 223
Web site 19, 71
website URL 237
Week Planner 200
WEP 75, 78
what's 128
What's New 10
Whisper Announce 200, 219
Wide Area 32, 37, 69, 149, 161, 165, 166
Wide Area Expansion 149
Wide Area Network 32, 69, 161, 166
Wide Area Networking Protocol 165
Width 80, 291
WiFi 21, 46, 72, 75, 76, 77, 78, 79, 97
wildcards 10
Wink-Start 142
WINS 164
Wire 18, 41, 46, 64, 65, 66, 72, 75, 78, 124, 128, 137, 172, 179, 190, 251
closet/switch 132
wire speed 137
Wired Equivalent Privacy 75, 78
Wireless 6, 18, 21, 22, 40, 42, 46, 68, 69, 71, 72, 75, 82, 97, 99, 128, 132, 172, 179, 181, 277, 294
Wireless Access Points 21, 72, 277
Wireless Module 294
Wireless Telephones 22, 46, 68, 71, 72, 75, 82, 97
Wireless VoIP 22
Within Compact Contact Center 253
Within SoftConsole 184
WLAN 72, 132
WLAN Compatibility List" 72
Word 237, 253
workflow 200
workforce 205
workgroup 79
workgroups 20, 253
World 89, 128, 161
Rest 230
WorldCom 142
worlds companies' LANs 128
World-Wide Source
Crystal Training 253
Worst Case 132, 291
Wrap Up 176, 287
WS-X4148-RJ45V 132
WS-X6348-RJ45V 132

- X -

X.21 149, 161, 164, 277, 294
X.21/V35 WAN 32
X.25 165, 168
xDSL 164
XLS 253
XM24 40, 42, 97
XM24 DSS Unit 97

- Y -

Year 6
Yes 55, 56, 58, 60, 64, 65, 66, 180, 197, 219

- Z -

Zetafax
Networks 205
Zip Code/Postal Code 253

© 2008 AVAYA All rights reserved.

Performance figures and data quoted in this document are typical, and must be specifically confirmed in writing by Avaya before they become applicable to any particular order or contract. The company reserves the right to make alterations or amendments to the detailed specifications at its discretion. The publication of information in this document does not imply freedom from patent or other protective rights of Avaya or others.

Intellectual property related to this product (including trademarks) and registered to Lucent Technologies have been transferred or licensed to Avaya.

All trademarks identified by the ® or ™ are registered trademarks or trademarks, respectively, of Avaya Inc. All other trademarks are the property of their respective owners.

This document contains proprietary information of Avaya and is not to be disclosed or used except in accordance with applicable agreements.

Any comments or suggestions regarding this document should be sent to "wgctechpubs@avaya.com".

© 2008 Avaya Inc. All rights reserved.

Avaya
Unit 1, Sterling Court
15 - 21 Mundells
Welwyn Garden City
Hertfordshire
AL7 1LZ
England

Tel: +44 (0) 1707 392200
Fax: +44 (0) 1707 376933

Web: http://marketingtools.avaya.com/knowledgebase